

Guide for BTIP and Business Talk SIP services Microsoft Teams Direct Routing

8 november 2023

- Teams Direct Routing AudioCodes Checklist 0.51
 - AudioCodes Analog Phones Checklist 1.1
 - AudioCodes FAX Checklist 1.2
- Teams Direct Routing Ribbon Checklist 1.14
 - Ribbon FAX Checklist 1.0
- Teams Direct Routing Oracle Checklist 0.21



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1 Main certified architectures

1.1 Teams media capabilities

1.1.1 Media Bypass

Teams Media Bypass has been certified so far with AudioCodes, Ribbon and Oracle SBC. As a reminder, Media Bypass means bypassing Teams cloud (actually Teams Media Processors) for media flows.

1.1.2 Local Media Optimization “Europe” model

Local Media Optimization “Europe” model has been certified with AudioCodes and Ribbon so far. With this feature, offnet media flows to/from internal users remain within BVPN or LAN, but are anchored to private LAN interface of DR SBC (see “Example 3 – offnet call from a BVPN remote site with Local Media Optimization Europe Model” below).

1.1.3 Local Media Optimization “Asia” model

➤ General overview

Local Media Optimization “Asia” model has been certified with AudioCodes so far, involving Proxy and Downstream SBC. It may be useful to keep media flow geographically local for users that are on very remote BVPN sites that may take advantage of a local Business Talk aSBC.

➤ BTIP in DROM (French Overseas Territories)

Despite its name, “Asia” model is applicable anywhere, especially in a “BTIP DROM” scope, where a central BTIP trunk is connected to a central Direct Routing SBC in European France, while users in French overseas territories take advantage of a local BTIP trunk connected to a local downstream SBC (see also 1.7 BTIP in DROM (French Overseas Territories)).

1.2 FAX

Concerning the FAX support, Business Talk and BTIP support the following usage:

- FAX servers connected to the IPBX* -and sharing same dial plan-,
- FAX servers as separate ecosystems -and separate dial plan-,
- Analog FAX machines, usually connected to specific gateways* (seen as IPBX ecosystem or not).

Note that FAX communications via Business Talk (International) will still be allowed but no longer officially supported by the Orange support teams from April 2023 for new customers implementations.

Microsoft Teams does not directly handle FAX, neither analog device. Those devices can only be interconnected with Teams through the Direct Routing SBC or ATA boxes linked to the Direct Routing SBC and would be fully managed by the Direct Routing SBC.

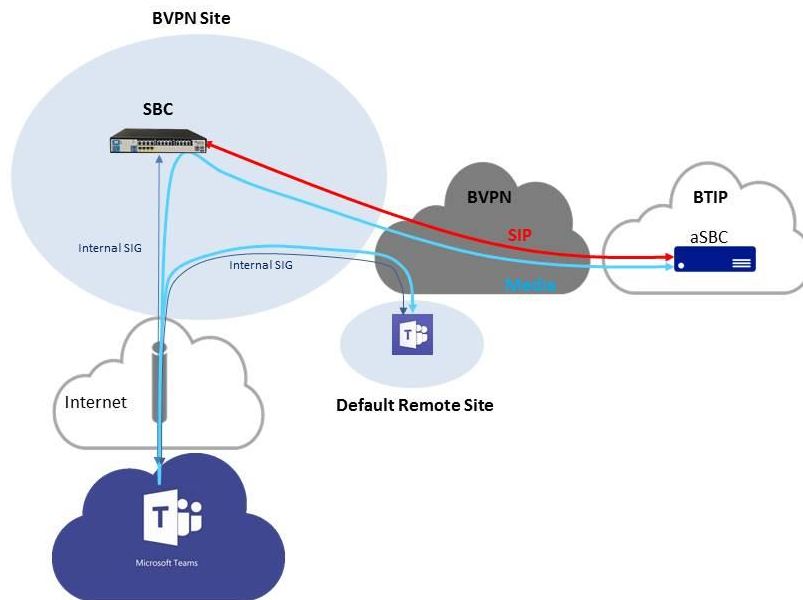
FAX flows are handled via T.38 transport only.

1.3 Redundancy

Note also that BTIP/BTalk/BToI/BTIPoI redundancy mechanisms are not shown on the drawings. This is not the aim of this document.

1.4 Standalone mode

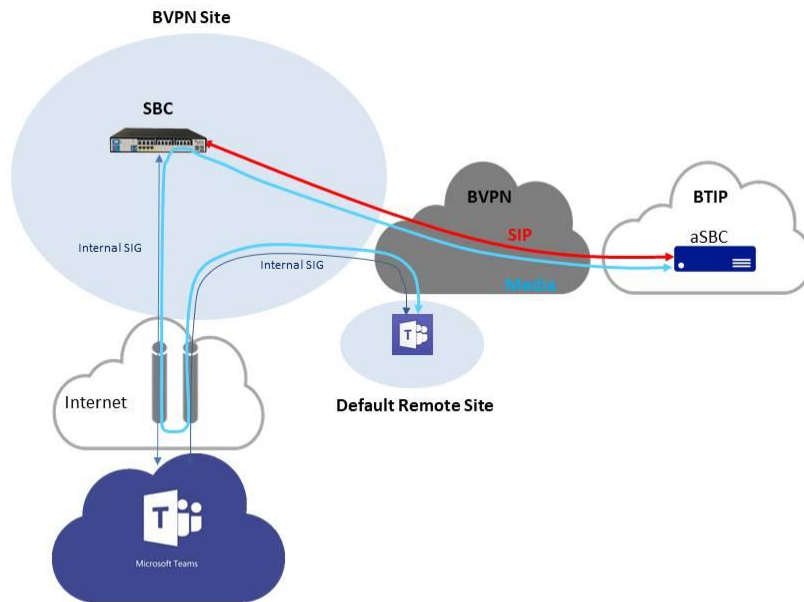
Example 1 – offnet call from a BVPN remote site without Media Bypass



Here, the Teams user belongs to a BVPN site, as does the SBC. There is no Internet breakout within the local site, though that could be the recommended architecture.

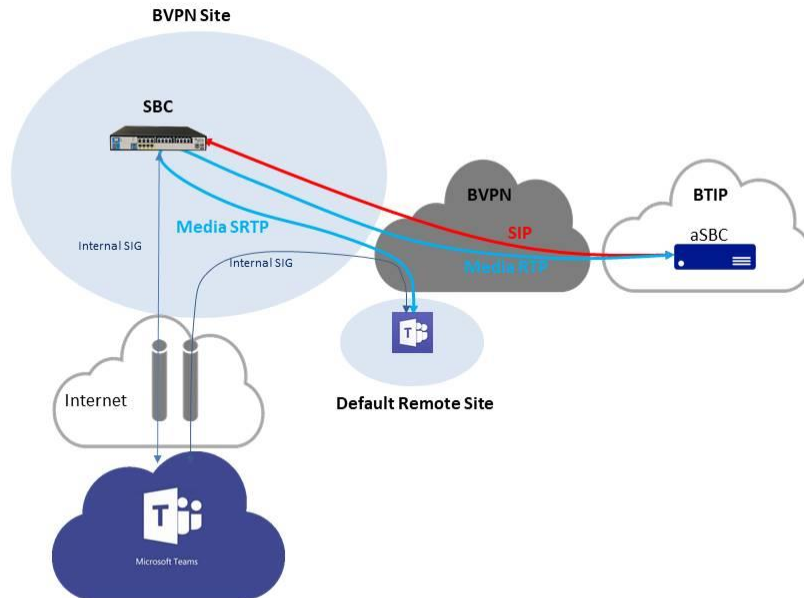
Without Media Bypass, media flows cross Internet to reach Teams Media Processors and SBC.

Example 2 – offnet call from a BVPN remote site with Media Bypass



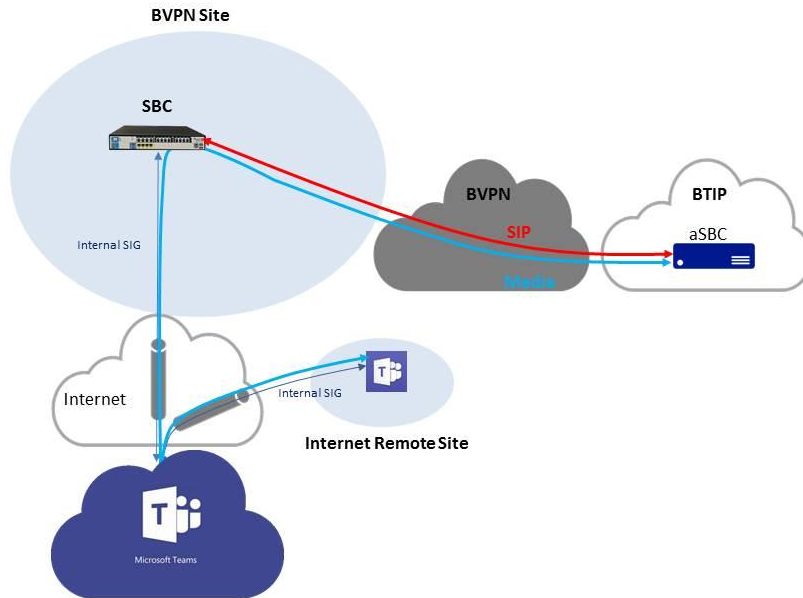
With Media Bypass, media flows cross Internet and join directly Teams user and SBC. They may cross a Microsoft Transport Relay for NAT traversal reasons, but Transport Relays are much more numerous than Media Processors and provide no time-consuming treatment of the media.

Example 3 – offnet call from a BVPN remote site with Local Media Optimization Europe Model

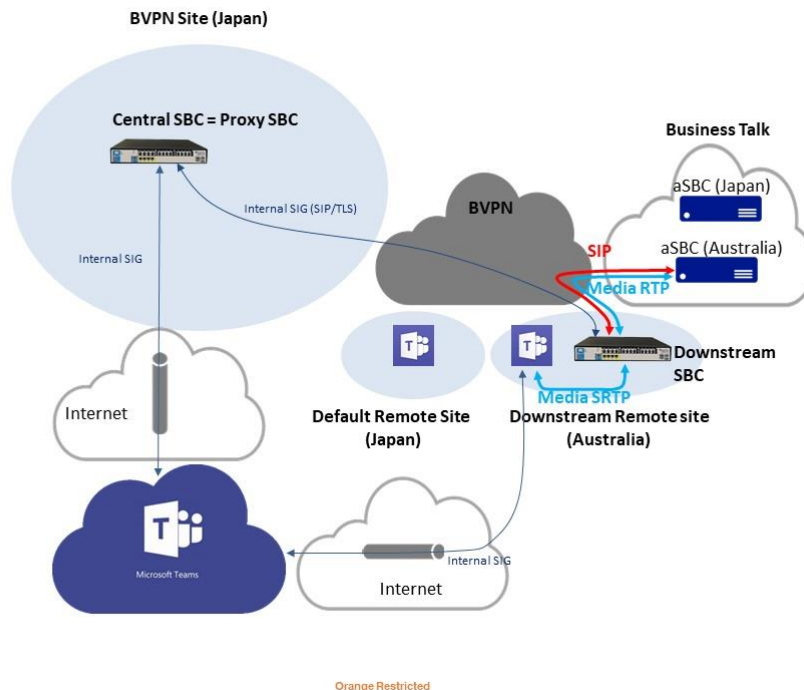


With Local Media Optimization configured, offnet media flow remains on BVPN or LAN for internal Teams users. It aims at using a shorter path to the BT/BTIP SIP trunk and better voice quality than Internet quality. It also saves Internet bandwidth. With “Europe” model, media flows are anchored to the private interface of the Direct Routing SBC. “Asia” model, with which direct media is expected, is not available yet with BT/BTIP.

Example 4 – offnet call from an Internet remote site without Media Bypass (and with or without Local Media Optimization)



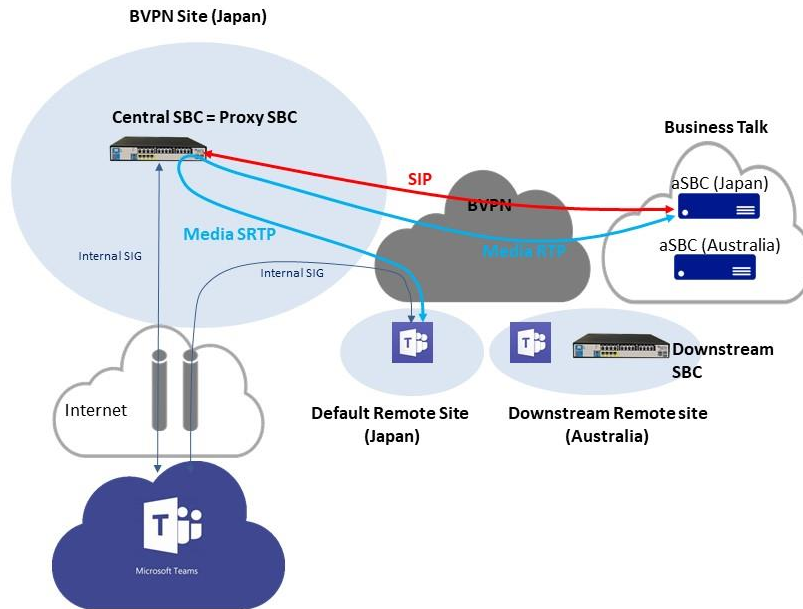
Here, the Teams user does not belong to a BVPN site and is located on Internet. Media Bypass is not activated. Local Media Optimization has no effect on call flows involving Internet users.

Example 5a – offnet call from a BVPN remote site with Local Media Optimization Asia model

Local Media Optimization Asia Model may be useful to keep media flow geographically local for users that are on very remote BVPN sites that may take advantage of a local Business Talk aSBC. For instance, the central SBC may be in Japan, while there are some users in Australia (drawing above). Another instance would be in a BTIP “DROM” context, with a central SBC in France and users in West French Indies. A valid solution would be to use locally a separate Direct Routing SBC, but Local Media Optimization Asia model allows to use local **downstream** SBC that does not need to have a public @IP because they need a trunk to the central SBC only. That avoids deploying local DMZ as no Internet access is required.

In this architecture, central SBC may be named **proxy** SBC as well, meaning it is the front-end SBC from Teams point of view, screening the **downstream** SBC.

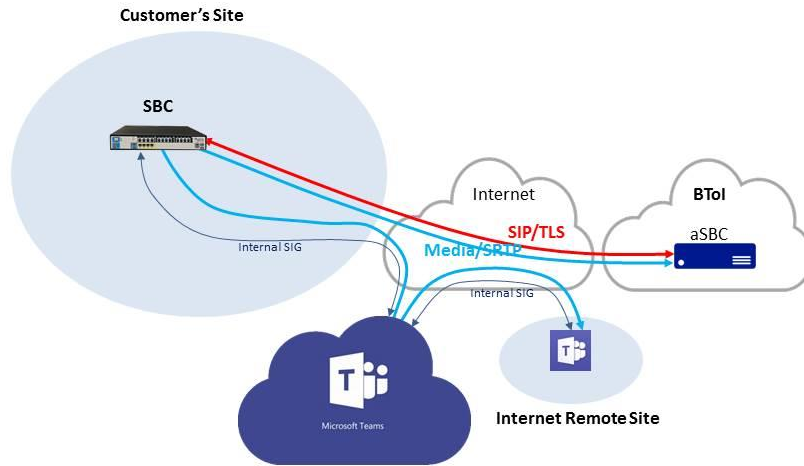
Example 5b – offnet call from a BVPN central site with Local Media Optimization Asia model



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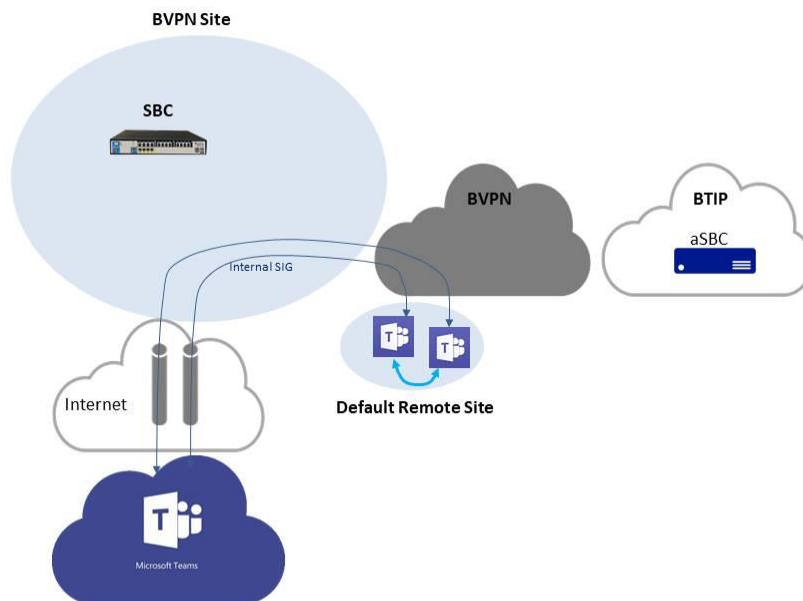
This example is linked to example 5a and shows that a user in Japan uses the central SBC that is trunked to the Japanese aSBC of Business Talk. This is like Local Media Optimization Europ model.

Example 6 – offnet call from an Internet remote site with BTol (Business Talk over Internet)



BVPN is not mandatory in a Business Talk over Internet architecture. Here is a full Internet architecture. Note that the flows are encrypted from end to end. Media Bypass is not activated, but it could be.

Example 7 – onnet BVPN call



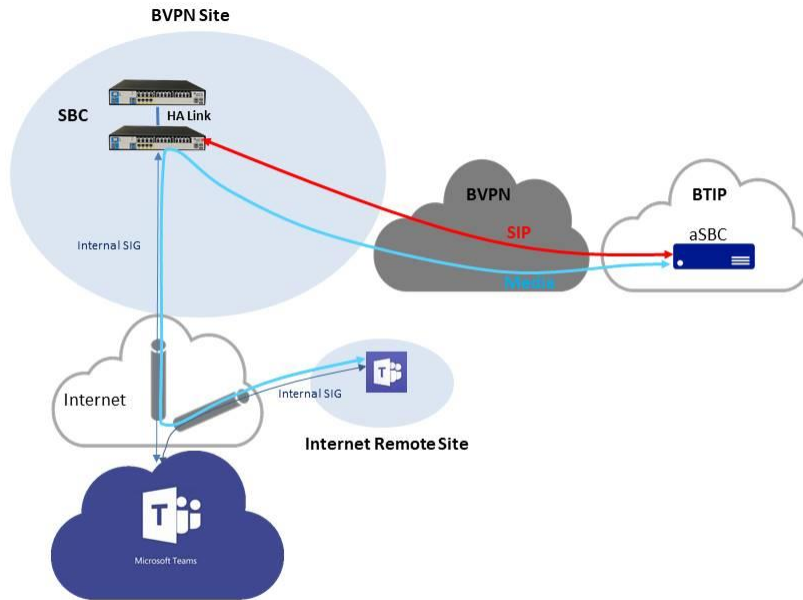
In a BVPN architecture, media flows between Teams users belonging to a single BVPN are direct, whether they are on the same site or in distinct sites.

1.5 High Availability mode

High availability is provided by SBC vendor. This is active-passive mode with a single IP address.

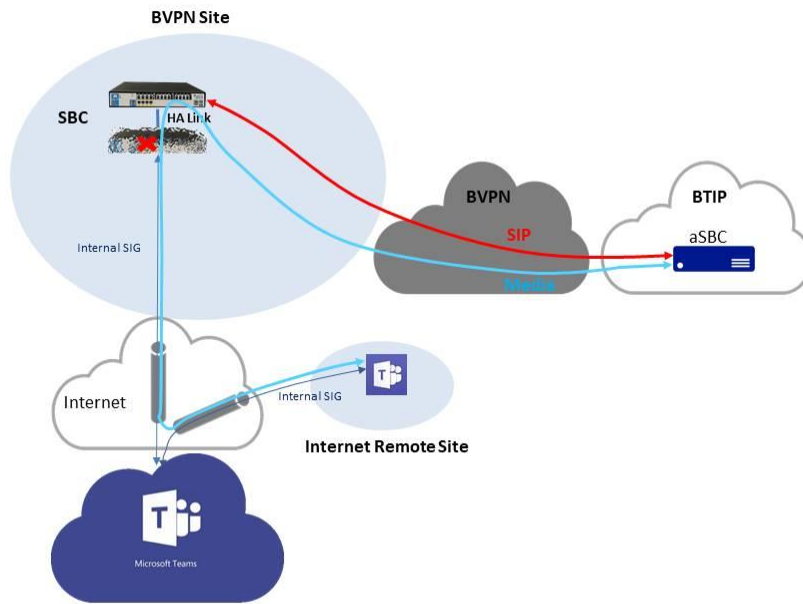
From BTIP/BTalk perspective, this is like a single SBC.

Example 1 – offnet call in nominal mode (and Media Bypass)



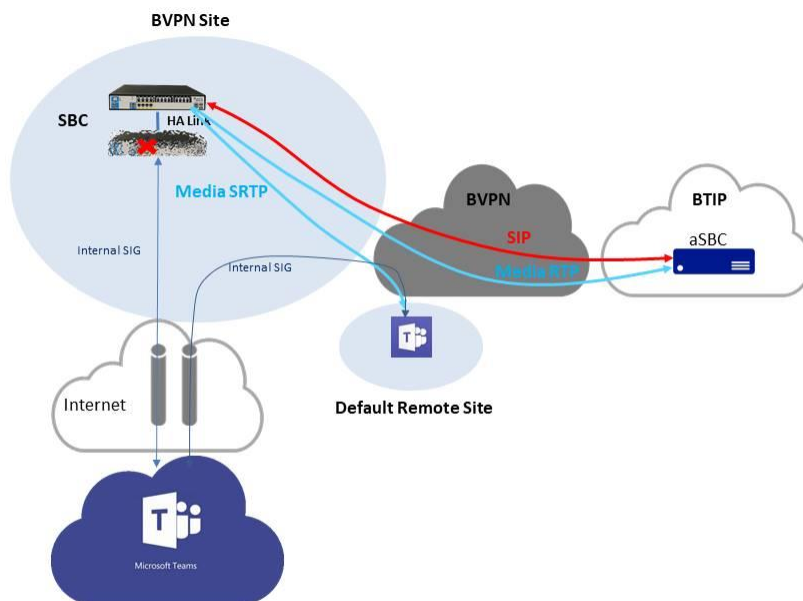
HA is fully managed by SBC themselves. From Teams and BTIP/BTalk points of view, the architecture behaves as if there was a single SBC.

Example 2 – offnet call in backup mode (and Media Bypass)



When the nominal SBC fails, the backup SBC transparently handles the service. Current calls are not cut. Here, Media Bypass is activated, but could not be.

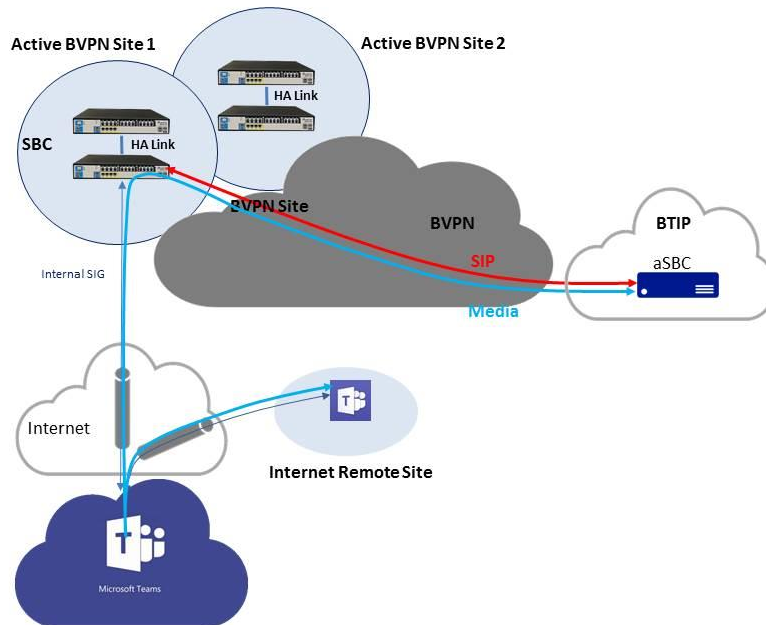
Example 3 – same as example 2 with a BVPN Teams user and Local Media Optimization “Europe” model



1.6 Active-active mode (resiliency)

From BTIP/BTalk perspective, meaning of active-active mode for incoming offnet calls is actually round-robin mode.

Example 1 – offnet call in an architecture combining two BVPN sites in active-active mode with two SBC in High Availability mode in each



Here is an example of a somewhat highly resilient architecture that mixes spatial redundancy with load balancing and high availability on each SBC site. The two geographic SBC sites are supposed to be both located in the same region as the a-SBC pair they are trunked to.

Teams use DNS load balancing between the two geographic SBC sites for routing outbound offnet calls. On the other and, BTIP/BTalk provides round-robin between the two sites. The two sites are active-active. In addition, there is here an active-passive resilience within each site.

In the next chapter describing connections to BTIP/BTalk, this architecture would be “N SBC pairs - BTalk round robin mode over N pairs in vendor HA mode” with N=2. Of course, a simpler architecture without the High Availability part could be considered as well.



1.7 BTIP in DROM (French Overseas Territories)

1.7.1 Definition

BTIP DROM is part of the french offer dedicated to multi-site companies with some locations in French Overseas departments (aka "DROM").

- IPBX located in continental France or in DROM
- Users located in continental France and in DROM

BTIP in DROM offer aims at keeping the media flows locally, typically for French customers with a central site in Continental France et other sites in DROM such as Caribbean or Indian Ocean. As a cloud service, Teams has no real location, but Direct Routing SBC may have a location. Flows may become more complex than with a regular IPBX and the way to optimizing them may sound sometimes not natural. For instance, as shown below, it is better not to have multiple trunks in case of a single Direct Routing SBC.

1.7.2 Choice of outbound trunk

In BTIP DROM, outbound trunk must depend on the calling site. A call initiated in a Guadeloupean site must be routed to the Caribbean aSBC. Local aSBC have been installed in following regions:

- Caribbean area: for sites of Martinique (+596), Guadeloupe, Saint Barthelemy, Saint Martin (+590), Guyane (+594)
- Réunion: for sites of La Réunion or Mayotte (+262)

SIP trunks must not be used when users call each other with their DID. The call must be considered as local. This is Teams standard behavior.

With Teams, this outbound constraint, and thus the use of BTIP DROM, may not be relevant in some architectures. For instance, as shown below, it is better not to have multiple trunks in case of a single Direct Routing SBC.

1.7.3 « To » header: supported formats for outbound calls to PSTN

Domestic number

- E164 format is preferred: +Country code followed by phone number without "0" prefix
Ex : +590 590 88 67 21, +262 262 25 79 02 ou +33 1 25 58 58 58
- Authorized: 0ZABPQMCDU ou 0262PQMCDU for DROM (ex: La Réunion)
Ex : 0 590 88 67 21, 0 262 25 79 02 ou 02 56 48 58 58
- Tolerated : +33 262PQMCDU for (ex: La Réunion)
Ex : +33 590 88 67 21 (La Guadeloupe) or +33 262 25 79 02

International number

- E164 : +CC NSN
Ex : +48 504135755

Short number

- +33 followed by short number (preferred) or short number
Ex : +33 3000 ou 3000

Emergency number

- +33 followed by emergency number (preferred) or emergency number
Ex : +33112 ou 112

1.7.4 « From » header: supported formats supportés for outgoing calls to PSTN

- E164 format is preferred: +Country code followed by phone number without "0" prefix
Ex : +33 3 68 45 57 56, +262 262 69 97 03 or +590 590 68 22 41
- Authorized: 0ZABPQMCDU or 0262PQMCDU for DROM (ex: La Réunion)
Ex: 03 68 45 57 56, 0 262 69 97 03 ou 0 590 68 22 41
- Tolerated: +33 262PQMCDU for DROM (ex: La Réunion)
Ex : +33 262 69 97 03 or +33 590 68 22 41 (La Guadeloupe)

1.7.5 Comparison of architectures with or without BTIP DROM

This chapter describes some possible architectures with Teams Direct Routing and the impact on voice quality for companies with sites in continental France and in DROM.

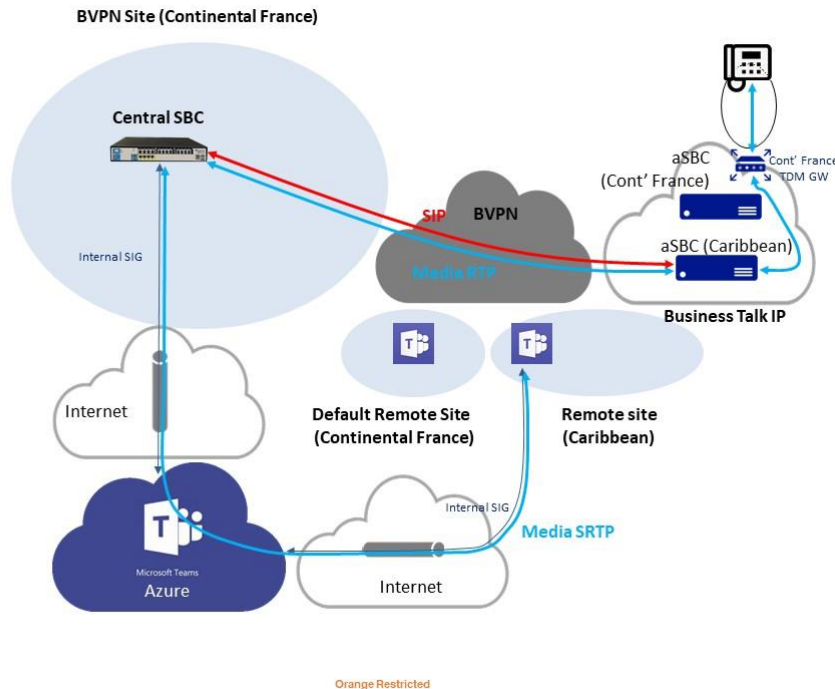
| | noMBP | MBP | LMOE | LMOA |
|---|--|-----------------|---------------------------------|------|
| Two trunks on central DR SBC, one with aSBC in Continental France and one with aSBC in DROM (1.7.6) | Not recommended | | | N/A |
| Single trunk on central DR SBC with aSBC in Continental France (1.7.7) | OK for administrative telephony | Not recommended | OK for administrative telephony | N/A |
| | This is standard BTIP and not BTIP DROM architecture | | | |
| One trunk per plate and one DR SBC per plate (1.7.8) | OK | | | N/A |
| One trunk per plate and LMO "Asia" with one downstream SBC per plate (1.7.9) | N/A | | | OK |

1.7.6 Two trunks on central DR SBC, one with aSBC in Continental France and one with aSBC in DROM

The aim of this architecture would be to take advantage of the DROM aSBC for local users. However, it is **not recommended** as shown in the following example, an offnet call between a Caribbean user and an offnet customer located in Continental France. In this case, 3 long legs are involved, leading to a possible bad voice quality:

- leg between Caribbean user and DR SBC located in Continental France,
- leg between DR SBC and DROM aSBC located in the Caribbean,

- leg between DROM aSBC in the Caribbean and TDM Gateway located in Continental France.



Note that the issue would be the same if Local Media Optimization “Europe” model would be enabled. The only difference would be the path of the first leg, remaining in BVPN instead of Internet and Azure, but still crossing the Atlantic Ocean.

1.7.7 Single trunk on central DR SBC with aSBC in Continental France

➤ Without Media ByPass

This architecture is valid for administrative telephony, though advantage is not taken of the DROM aSBC. No Local Media Optimization is enabled in Microsoft Teams Direct Routing. The worst case would be considering an offnet call between a Caribbean user and a local customer. 2 long legs are involved:

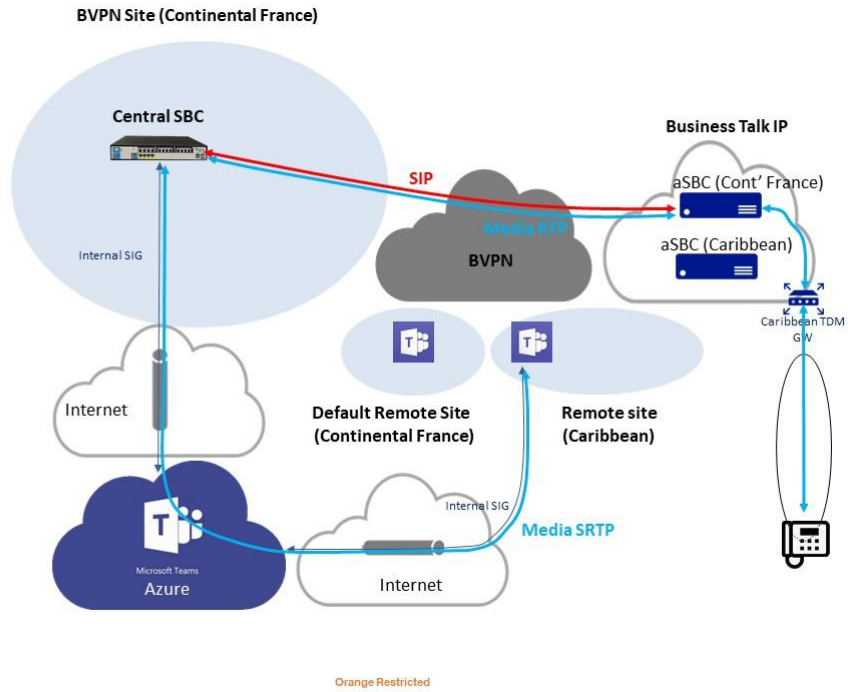
- between Caribbean user and DR SBC located in Continental France, through Internet and Azure (assuming Media Bypass is not enabled),
- between BTIP aSBC in Continental France and Caribbean TDM Gateway, through BVPN.

The highest measured Round-Trip Delay (RTD) in BVPN between Continental France and DROM leads to a MOS > 4 (few users dissatisfied). It concerns the Indian Ocean and is lower for the Caribbean.



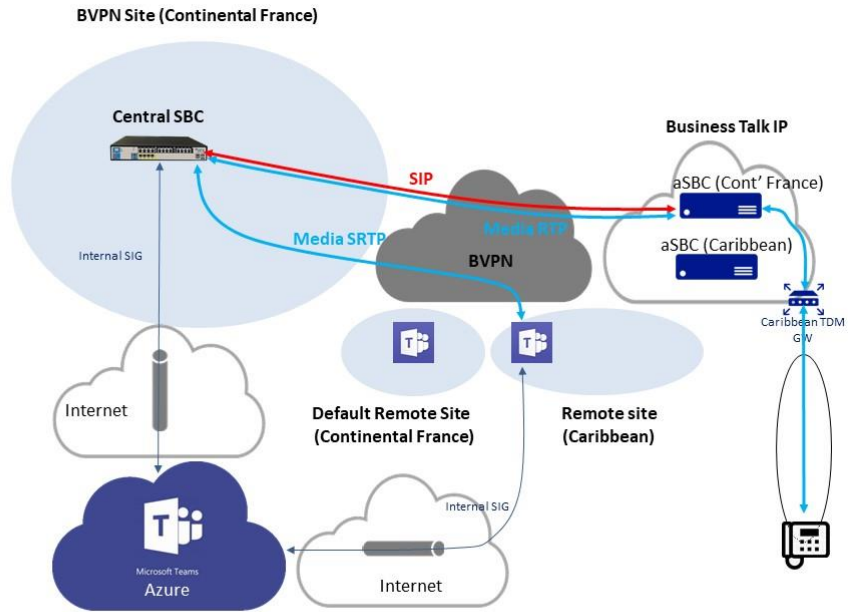
However, here, half on the flow only is on BVPN. The other part uses Internet and Azure. Microsoft networking ensures that most part of the leg stays in Azure network, which implements Quality of Service, and a little is over the Internet. That may be enough for administrative telephony.

With Media Bypass, the part of the Internet may be higher. This is why it is less recommended.



➤ **with Local Media Optimization “Europe”**

With LMO “Europe”, there is no leg on Internet + Azure and only the Round-Trip Delay on BVPN is relevant. That may be enough for administrative telephony as well.



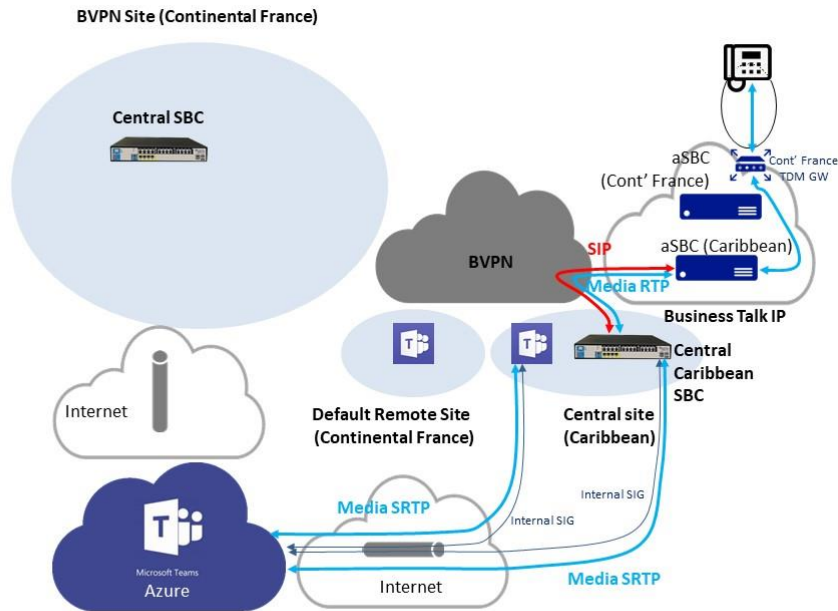
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1.7.8 One trunk per plate and one DR SBC per plate

➤ **Without Media ByPass**

This architecture is basically deploying two Direct Routing SBC on the same tenant, each one owning its BTIP trunk, here without Media Bypass. Offnet calls are routed locally.

Worst case would be calls between a Caribbean user and a Continental French customer (or reversely), that cross the Atlantic Ocean only once through BVPN between Caribbean aSBC and Continental French TDM Gateway. This architecture is recommended for good voice quality.

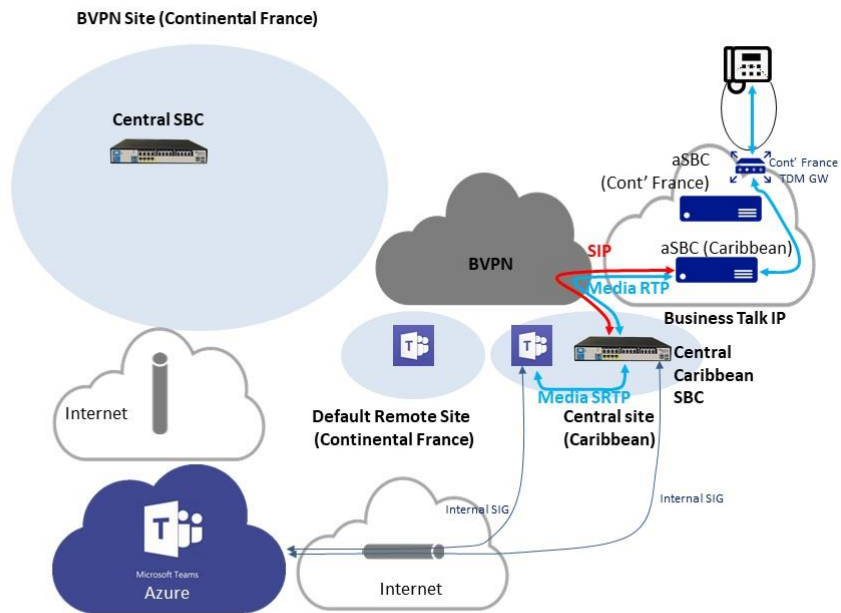


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➤ With Local Media Optimization “Europe”

This architecture is a variant of above. It is basically deploying two Direct Routing SBC on the same tenant, each one owning its BTIP trunk, with Local Media Optimization “Europe” model enabled. Offnet calls are fully routed locally.

Worst case would be calls between a Caribbean user and a Continental French customer (or reversely), that cross the Atlantic Ocean only once through BVPN between Caribbean aSBC and Continental French TDM Gateway. This architecture is fully recommended for good voice quality.

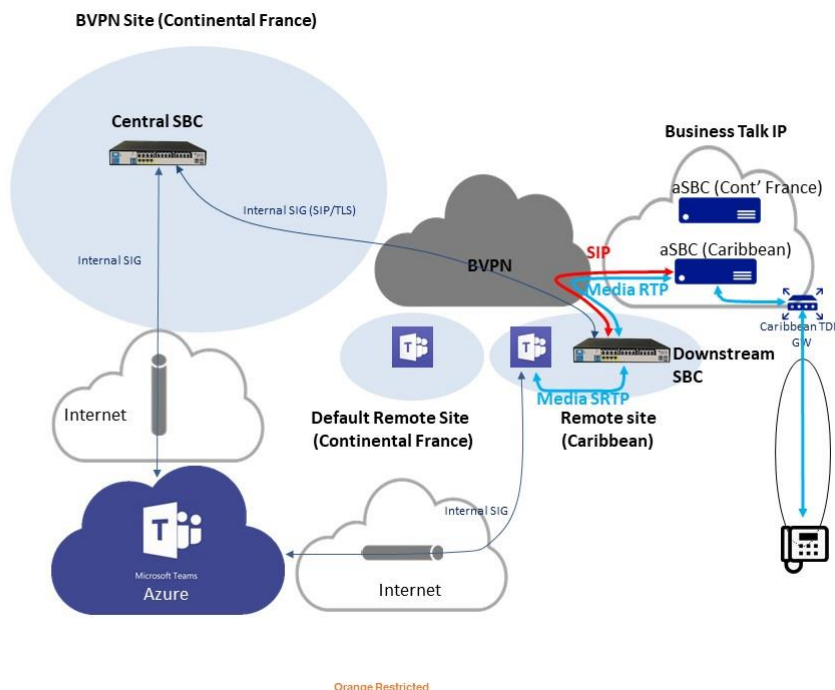


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1.7.9 One trunk per plate and LMO “Asia” with one downstream SBC per plate

The architecture depicted here uses Local Media Optimization “Asia” model already described at 1.4. The main difference with above is that the Downstream SBC has no trunk with Teams, but only with the Central SBC and therefore does not need to be open on the Internet.

This is also a recommended architecture for BTIP DROM.



1.7.10 Operator Connect

TBD

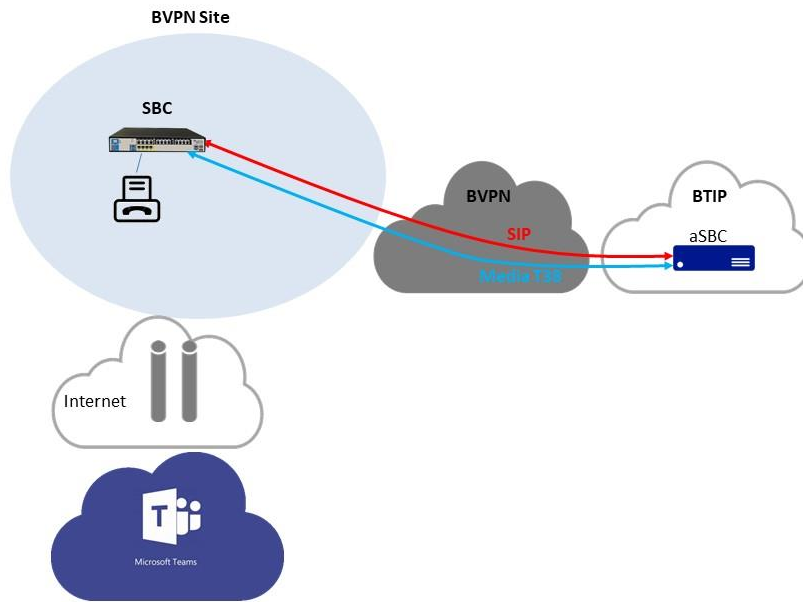
1.8 Analog devices

1.8.1 FAX

FAX on Direct Routing SBC with or without Gateway is certified both on French (BTIP) and International (BTalk) scopes. FAX protocol is T.38.

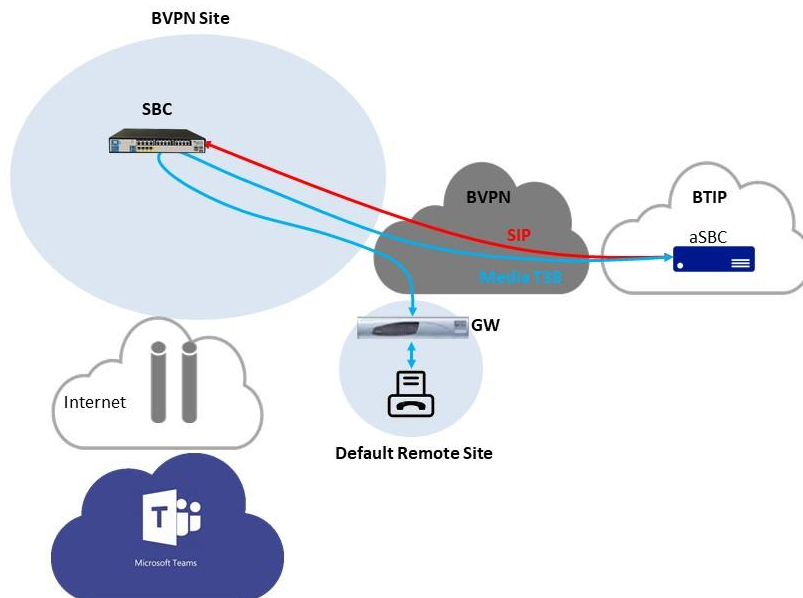
FAX calls to and from Business Talk consumes the same SIP Trunk that is used for interconnection with Teams. When a call is made from FAX or to FAX, the Direct Routing SBC bypasses Teams, which does not talk T.38, for signaling as well as for media

Example 1 - FAX directly connected to SBC



The analog FAX device can be connected directly to hardware DR SBC with FXS ports. Call is routed directly between Business Talk / Business Talk IP and FAX.

Example 2 - FAX connected to a cascaded GW behind SBC



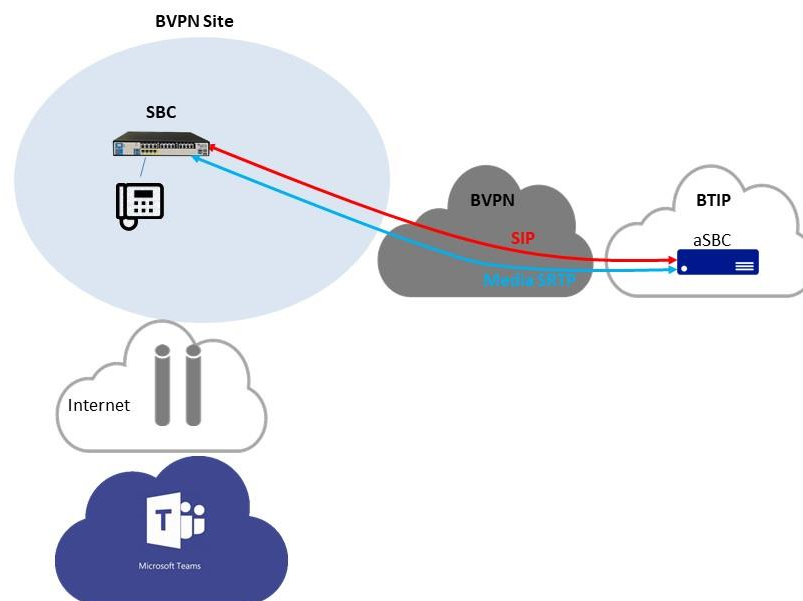
In this architecture FAX device is connected to an analog telephony adapter. It is integrated with DR SBC which can be placed in other remote site or in datacenter. DR SBC with no directly connected endpoints can be virtualized.

Same as in previous architecture call is routed directly between Business Talk / Business Talk IP and FAX and bypasses Teams. This has been certified so far with AudioCodes central SBC with AudioCodes GW (MPxxx or Mediant) and Ribbon Edge central SBC with Ribbon GW 1000/2000.

1.8.2 Analog phones

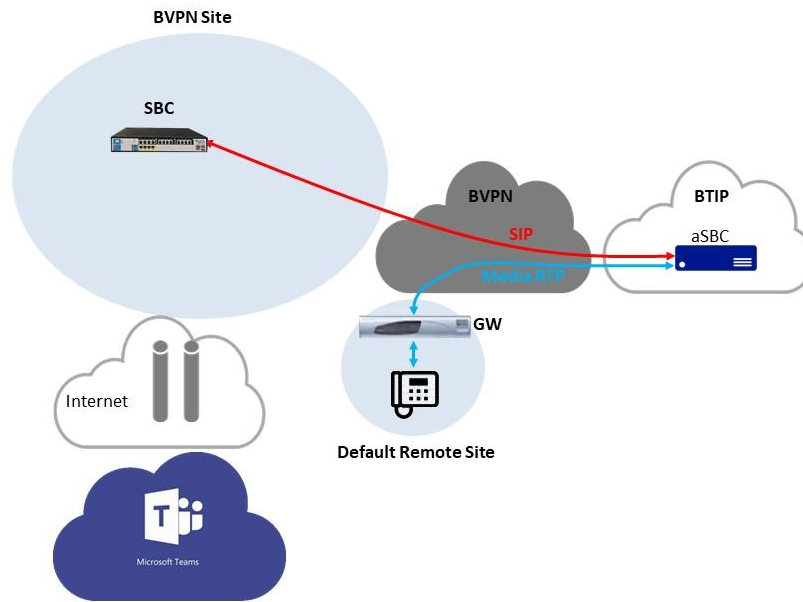
Analog phones on Direct Routing SBC with or without Gateway is certified both on French (BTIP) and International (BTalk) scopes, so far with AudioCodes only as central SBC and remote GW (MPxxx or Mediant).

Example 1 – Analog phone directly connected to central SBC



The analog phone can be connected directly to hardware DR SBC with FXS ports. Call is routed directly between Business Talk / Business Talk IP and analog device.

Example 2 – Analog phone connected to a cascaded GW behind SBC



In this architecture, device is connected to an analog telephony adapter. It is integrated with DR SBC which can be placed in other remote site or in datacenter. DR SBC with no directly connected endpoints can be virtualized. Note that direct media with BTIP/BTalk has been validated in this case.

1.9 Reminder on emergency calls

(Extract of Access SIP profile for connecting PBX to BTIP/BT SIP or FIAs2 services V3.5)

PBX may send a PIDF-LO body for providing geographical information BUT this information must be fulfilled with great care because, if provided, this will be used in case of Emergency call for locating caller. OBS is not responsible for the relevance of the provided information or its format.

Generally, PIDF-LO must not be used on BTIP (French trunk offer). In any case, please contact OBS before using this PIDF-LO body.

1.10 Sizing considerations

1.10.1 Coders

Regarding codec usage in the standard architecture:

RTP G711, G729 or G722 is used between Direct Routing SBC and BTIP/BT (see 3.3 for more information on supported codecs on trunk),

Same codec is used in SRTP between Teams and The Direct Routing SBC so that transcoding is avoided, SRTP overhead compared to RTP is neglectable,

encrypted SILK or G722 is used between Teams Client and Teams Media Processors,

Therefore, a channel for an offnet call may be considered to consume at most 100 kbps throughout its path.

1.10.2 Real Time Voice (RTVo) classification

In Business VPN, voice flows are classified either by using “Access Control Lists” on CE routers or by trusting DSCP configuration of voice endpoints. “DSCP trust” is intended to become the main way of managing QoS. Therefore, take care to have the following DSCP values configured on your equipment (SBC, GW, Teams client, hardphones...):

- Voice media: **46** (= EF) *!! mandatory !!*
- Video media: **26** (=AF31) or **34** (= AF41)
- Signaling: : **24** (=CS3) or **26** (=AF31) or **40** (= CS5) or **46** (= EF)

Note that our configuration guidelines below include this configuration for SBC and Teams client.

Teams proposes no CAC mechanism to ensure that the RTVo bandwidth won't be overloaded and that may lead to packet loss in case of BVPN access saturation. It's up to you to take care of you consumption and properly size your access with the help of your sales contact.



1.10.3 Table of used channels

Following table recaps the use of channels for selected call flows. For onnet calls, channels consumed by possibly involved far end site are not considered.

| | Central site | | Remote site with Internet breakout | | Remote site without Internet breakout | Remote site with downstream SBC (LMO Asia) |
|---|----------------|-----------------|------------------------------------|-----------------|---------------------------------------|--|
| | CE Router RTVo | Internet router | CE Router RTVo | Internet router | CE Router RTVo | CE Router RTVo |
| Offnet calls without Local Media Optimization | | | | | | |
| Client on central site | 1 | 2 | | | | |
| Client on remote site w Internet breakout | 1 | 1 | | 1 | | |
| Client on remote site wo Internet breakout | 2 | 2 | | | 1 | |
| Offnet calls with Local Media Optimization Europe | | | | | | |
| Client on central site | 1 | | | | | |
| Client on remote site w Internet breakout | 2 | | 1 | | 1 | |
| Client on remote site wo Internet breakout | 2 | | 1 | | 1 | |
| Offnet calls with Local Media Optimization Asia | | | | | | |
| <i>Client served by central SBC</i> | | | | | | |
| Client on central site | 1 | | | | | |
| Client on remote site w Internet breakout | 2 | | 1 | | 1 | |
| Client on remote site wo Internet breakout | 2 | | 1 | | 1 | |
| <i>Client served by downstream SBC</i> | | | | | | |
| Client on remote site with downstream SBC | | | | | | 1 |
| Client on remote site w Internet breakout | | | 1 | | 1 | 2 |
| Client on remote site wo Internet breakout | | | 1 | | 1 | 2 |
| Offnet calls from/to Analog devices | | | | | | |
| <i>Without Local Media Optimization Asia (device served by central SBC)</i> | | | | | | |
| FAX on central site | 1 | | | | | |
| FAX on remote site w Internet breakout | 2 | | 1 | | | |
| FAX on remote site wo Internet breakout | 2 | | | | 1 | |
| Analog Phone (AP) on central site | 1 | | | | | |
| AP on remote site w Internet breakout | | | 1 | | | |
| AP on remote site wo Internet breakout | | | | | 1 | |
| <i>With Local Media Optimization Asia (device served by downstream SBC)</i> | | | | | | |
| FAX on remote site with downstream SBC | | | | | | 1 |
| FAX on remote site w Internet breakout | | | 1 | | | 2 |
| FAX on remote site wo Internet breakout | | | | | 1 | 2 |
| Analog Phone (AP) on downstream SBC | | | | | | 1 |
| AP on remote site w Internet breakout | | | 1 | | | |
| AP on remote site wo Internet breakout | | | | | 1 | |
| Onnet BVPN calls | | | | | | |
| From/to a local client | | | | | | |
| Between central site and a remote client | 1 | | | | | |
| Between remote site w Internet breakout and a remote client | | | 1 | | | |
| Between remote site wo Internet breakout and a remote client | | | | | 1 | |
| Between AP on central site and a remote client | 1 | | | | | |
| Between AP on remote site w Internet breakout and a remote client | | | 1 | | | |
| Between AP on remote site w Internet breakout and a remote client | | | | | 1 | |
| External calls (between BVPN & Internet clients) | | | | | | |
| Client on central site | | 1 | | | | |
| Client on remote site w Internet breakout | | | | 1 | | |
| Client on remote site wo Internet breakout | 1 | 1 | | | 1 | |



2 Parameters for connection to BTIP/BTalk/BToI/BTIPoI

2.1 Trunking integration

| Head Quarter (HQ) architecture | Level of Service | @IP / FQDN used by the service |
|--|---|---|
| Single SBC | No redundancy | BTIP/BTalk : SBC IP@ BTol: SBC public IP@ or FQDN DNS Type A BTIPoI: SBC public FQDN type A |
| SBC pair - Vendor HA mode | Local redundancy with nominal/backup behavior. No loss of calls in case of nominal's failure. Single site. | BTIP/BTalk: SBC pair floating IP@ BTol: SBC public floating IP@ or FQDN DNS Type A BTIPoI: SBC public floating FQDN DNS Type A |
| | Extended local redundancy with nominal/backup behavior. No loss of calls in case of nominal's failure. Two sites linked at L2 level. | |
| N SBC - BTalk round robin mode - Mono region | Local redundancy and load sharing with round robin behavior. Loss of calls handled by a SBC that fails. DNS load balancing from Teams point of view. | BTIP/BTalk: SBC1 IP@ SBC2 IP@ ... SBCN IP@ BTol: SBC1 public IP@ or FQDN DNS Type A SBC2 public IP@ or FQDN DNS Type A ... SBCN public IP@ or FQDN DNS Type A BTIPoI: SBC1 public FQDN DNS Type A SBC2 public FQDN DNS Type A ... SBCN public FQDN DNS Type A |
| | Geographical redundancy and load sharing with round robin behavior. Loss of calls handled by a SBC that fails. DNS load balancing from Teams point of view. M sites (M <= N) in the same region (i.e. attached to the same aSBC pair with a single T1T7). | |



| | | |
|--|--|---|
| <p>N SBC pairs - BTalk round robin mode over N pairs in vendor HA mode - Mono region</p> | <p>Local + geographical redundancy and load sharing. If a SBC fails locally, active calls not lost and handled by new active SBC. If a full site fails, active calls are lost, but all users keep the offnet service. In addition, there is a load-balancing between sites : round-robin from BTIP/BT side and DNS load balancing from Teams side.</p> | <p><u>BTIP/BTalk:</u> SBC pair1 floating IP@ SBC pair2 floating IP@ ... SBC pairN floating IP@ <u>BTol:</u> SBC pair1 public floating IP@ or FQDN DNS Type A SBC pair2 public floating IP@ or FQDN DNS Type A ... SBC pairN public floating IP@ or FQDN DNS Type A <u>BTIPol:</u> SBC pair1 public floating FQDN DNS Type A SBC pair2 public floating FQDN DNS Type A ... SBC pairN public floating FQDN DNS Type A</p> |
| <p>N SBC – DNS Resiliency - Mono region</p> | <p>Local redundancy and load sharing with DNS resiliency behavior. Loss of calls handled by a SBC that fails. DNS load balancing from Teams point of view.</p> <hr/> <p>Geographical redundancy and load sharing with DNS resiliency behavior. Loss of calls handled by a SBC that fails. DNS load balancing from Teams point of view. M sites (M <= N) in the same region (i.e. attached to the same aSBC pair with a single T1T7).</p> | <p><u>BTIP/Talk:</u> Not applicable <u>BTol or BTIPol:</u> SBC public FQDN DNS Type SRV</p> |
| <p>N SBC pairs – DNS Resiliency over N pairs in vendor HA mode - Mono region</p> | <p>Local + geographical redundancy and load sharing. If a SBC fails locally, active calls not lost and handled by new active SBC. If a full site fails, active calls are lost, but all users keep the offnet service. In addition, there is a load-balancing between sites : DNS resiliency BTIPol/BTol side and DNS load balancing from Teams side</p> | <p><u>BTIP/BTalk:</u> Not applicable <u>BTol or BTIPol:</u> SBC pairs public FQDN DNS Type SRV</p> |

| | | |
|---|--|---|
| <p>2 SBC pairs - BTalk T1T7 rescue mode over 2 pairs in vendor HA mode - Multi region (Business Talk only).</p> <p>Limited to 2 pairs for easier understanding. May be extended to N pairs, at least one per involved region.</p> | <p>For users in region 1: offnet calls to/from users of the region 1 are routed through the Direct Routing SBC pair of region 1, located in site 1. For Business Talk, a T1T7 per region permits this routing. For Teams, a second route must be configured for users. If a single SBC in site 2 fails, the backup SBC handles the traffic. If a single SBC in site 1 fails, the backup SBC handles the traffic. If the full site fails, Direct Routing SBC pair of region 2 rescues the offnet service of users, assuming that RTD is acceptable for VoIP.</p> <p>For users in region 2: offnet calls to/from users of the region 2 are routed through the Direct Routing SBC pair of region 2, located in site 2. For Business Talk, a T1T7 per region permits this routing. For Teams, a second route must be configured for users. If a single SBC in site 2 fails, the backup SBC handles the traffic. If the full site fails, Direct Routing SBC pair of region 1 rescues the offnet service of users, , assuming that RTD is acceptable for VoIP.</p> | <p><u>BTIP/BTalk:</u> SBC pair1 floating IP@ SBC pair2 floating IP@</p> <p>BTol: SBC pair1 public floating IP@ or FQDN DNS Type A SBC pair2 public floating IP@ or FQDN DNS Type A</p> <p>BTIPol: SBC pair1 public floating FQDN DNS Type A SBC pair2 public floating FQDN DNS Type A</p> |
| <p>2 SBC Nominal / Backup mode</p> | <p>- Local redundancy: both SBC are hosted on the same site, though HA is preferred here if supported by SBC vendor OR - Geographical redundancy both SBC are hosted on 2 different sites with WAN connectivity</p> | <p><u>BTIP/BTalk:</u> SBC1 IP@ SBC2 IP@</p> <p>BTol: SBC1 public IP@ or FQDN DNS Type A SBC2 public IP@ or FQDN DNS Type A</p> <p>BTIPol: SBC pair1 public FQDN DNS Type A SBC pair2 public FQDN DNS Type A</p> |
| <p>2 SBC pairs Nominal / Backup mode</p> | <p>'- Local and geographical redundancy SBC pairs are hosted on 2 different sites with WAN connectivity, each pair on one site</p> | <p><u>BTIP/BTalk:</u> SBC pair1 IP@ SBC pair2 IP@</p> <p>BTol: SBC pair1 public IP@ or FQDN DNS Type A SBC pair2 public IP@ or FQDN DNS Type A</p> <p>BTIPol:</p> |



| | | |
|--|--|--|
| | | SBC pair1 public FQDN DNS Type A SBC pair2 public FQDN DNS Type A |
|--|--|--|

| Remote Site (RS) architecture** | Level of Service | |
|---|--|--|
| Remote site without SBC | No survivability, no trunk redundancy | |
| Remote site with single downstream SBC for Local Media Optimization "Asia model" | Allows offnet media for local users not to transit through central SBC, but stay local. No redundancy of the downstream SBC | BTIP DROM/BTalk : Downstream SBC IP@ BTol: Downstream SBC public IP@ or FQDN DNS Type A BTIPol DROM: Downstream SBC public FQDN type A |
| Remote site with downstream SBC pair for Local Media Optimization "Asia model" - Vendor HA mode | Local redundancy of the downstream SBC with nominal/backup behavior. No loss of calls in case of nominal's failure. Single site. | BTIP/BTalk: Downstream SBC pair floating IP@ BTol: Downstream SBC public floating IP@ or FQDN DNS Type A BTIPol: Downstream SBC public floating FQDN DNS Type A |
| | Extended local redundancy of the downstream SBC with nominal/backup behavior. No loss of calls in case of nominal's failure. Two sites linked at L2 level. | |
| Remote site with N downstream SBC for Local Media Optimization "Asia model" - BTalk round robin mode - Mono region (Premium architecture) | Local redundancy and load sharing of the downstream SBC with round robin behavior. Loss of calls handled by a SBC that fails. Single site. | BTIP/BTalk: Downstream SBC1 IP@ Downstream SBC2 IP@ ... Downstream SBCN IP@ BTol: Downstream SBC1 public IP@ or FQDN DNS Type A Downstream SBC2 public IP@ or FQDN DNS Type A ... Downstream SBCN public IP@ or FQDN DNS Type A BTIPol: Downstream SBC1 public FQDN DNS Type A Downstream SBC2 public FQDN DNS Type A ... Downstream SBCN public FQDN DNS Type A |
| | Geographical redundancy and load sharing of the downstream with round robin behavior. Loss of calls handled by a SBC that fails. M sites (M <= N) in the same region (i.e. attached to the same aSBC pair with a single T1T7). | |



| | | |
|--|--|--|
| <p>Remote site with N downstream SBC pairs for Local Media Optimization "Asia model" - BTalk round robin mode over N pairs in vendor HA mode - Mono region (Premium+ architecture)</p> | <p>Local + geographical redundancy and load sharing of the downstream SBC. If a SBC fails locally, active calls not lost and handled by new active SBC. If a full site fails, active calls are lost, but all users keep the offnet service. In addition, there is a load-balancing between sites: round-robin from BTIP/BT side.</p> | <p><u>BTIP/BTalk:</u> Downstream SBC pair1 floating IP@ Downstream SBC pair2 floating IP@ ... Downstream SBC pairN floating IP@</p> <p><u>BTol:</u> Downstream SBC pair1 public floating IP@ or FQDN DNS Type A Downstream SBC pair2 public floating IP@ or FQDN DNS Type A ... Downstream SBC pairN public floating IP@ or FQDN DNS Type A</p> <p><u>BTIPol:</u> Downstream SBC pair1 public floating FQDN DNS Type A Downstream SBC pair2 public floating FQDN DNS Type A ... Downstream SBC pairN public floating FQDN DNS Type A</p> |
| <p>Remote site with 2 downstream SBC for Local Media Optimization "Asia model" Nominal / Backup mode</p> | <p>- Local redundancy: both downstream SBC are hosted on the same site, though HA is preferred here if supported by SBC vendor OR - Geographical redundancy both SBC are hosted on 2 different sites with WAN connectivity</p> | <p><u>BTIP/BTalk:</u> Downstream SBC1 IP@ Downstream SBC2 IP@</p> <p><u>BTol:</u> Downstream SBC1 public IP@ or FQDN DNS Type A Downstream SBC2 public IP@ or FQDN DNS Type A</p> <p><u>BTIPol:</u> Downstream SBC pair1 public FQDN DNS Type A Downstream SBC pair2 public FQDN DNS Type A</p> |
| <p>Remote site with 2 downstream SBC pairs for Local Media Optimization "Asia model" Nominal / Backup mode</p> | <p>- Local and geographical redundancy Downstream SBC pairs are hosted on 2 different sites with WAN onnectivity, each pair on one site</p> | <p><u>BTIP/BTalk:</u> Downstream SBC pair1 IP@ Downstream SBC pair2 IP@</p> <p><u>BTol:</u> Downstream SBC pair1 public IP@ or FQDN DNS Type A</p> |

| | | |
|--|--|---|
| | | Downstream SBC pair2 public IP@ or FQDN DNS Type A BTIPol: Downstream SBC pair1 public FQDN DNS Type A Downstream SBC pair2 public FQDN DNS Type A |
|--|--|---|

2.2 SBC IP addressing requirements

For BT/BTIP:

- **Internet facing Interface:** 1 Public IP address + 1 Private IP address NATed to the Public IP
 - Note: cf. NAT translation rule.
- **BT/BTIP SIP trunk Interface:** 1 IP address
- **Management Interface:** 1 IP address
 - Note: For BT/BTIP customers, this interface is generally shared with the BT/BTIP SIP trunk interface. This assumption has been taken in this document.
- **[Optional] HA interface:** 1 IP address (non-routed)

For Business Talk over Internet (BTol), or BTIPol :

- **Internet facing Interface:** 1 Public IP

Notes:

 - For BTol/BTIPol customers, this interface is shared between TEAMS and BTol/BTIPol SIP Trunk.
 - BTol/BTIPol recommends not to use NAT (cf. NAT translation rule).
- **Management Interface:** 1 IP address
- **[Optional] HA interface:** 1 IP address (non-routed)

2.3 ACL for BTol/BTIPol

The WAN or public IP interface is usually exposed to the public Internet through a DMZ, so it is strongly recommended to use an Access Control List for protecting access.

For instance, as a prerequisite, Ribbon recommends reading the [SBC Edge Security Hardening Checklist](#) to understand how to secure the SBC into your network infrastructure.

2.4 TLS integration for BTol/BTIPol

Root and intermediate Certificate (PEM format) must be transmitted to Orange BTIP/BTalk team.



3 BTIP/BTalk/BToI/BTIPol certified versions

3.1 Teams

There is no release numbering of Microsoft Teams. This is continuous delivery. Therefore, the compatibility of the 3 main components, i.e. certified DR SBC w Teams and BTIP/BTalk cannot be fully committed at any time.

3.2 Certified SBC

Certified with standard architectures are:

| Vendor | Model | Release | BTIP | BTalk | BToI | BTIPol ⁽¹⁾ | Codec | MBP ⁽²⁾ | LMO ⁽³⁾ | AP ⁽⁴⁾ | FAX |
|-------------|---|-------------------------|------|-------|---------------|-----------------------|-------------------------|--------------------|---------------------|-------------------|-----|
| Audio-Codes | M500/800/1000/2600/4000/9000 & VE | v.7.20A, build 258.475+ | ✓ | ✓ | ✓ | ✓ | G.722 G.711 G.729 | ✓ | ✓ Europe Asia | ✓ | ✓ |
| Ribbon | Edge 1000/2000 & SweLite | 9.0.4+ (but not 9.0.5) | ✓ | ✓ | ✓ | ✓ | G.711 Other planned | ✓ | ✓ Europe | Plan-ned | ✓ |
| Oracle | Acme Packet 1100/3900/4600 /6300/6350 & VME | S-Cz8.4.0 p2+ | ✓ | ✓ | Cust. request | Cust. request | G.722 G711 G729 | ✓ | Ongoing | No | No |

(1) BTIPol (BTIP Internet) service is not yet open at the time this document is produced. Please check with your sales contact

(2) MBP = Media Bypass

(3) LMO = Local Media Optimization "Europe" model and/or "Asia" model

(4) AP = Analog Phones

3.3 Codecs on trunk

3.3.1 Monocodec

Only G711 (A or μ laws) and G.729 are allowed

3.3.2 Multicodec (provided that Multicodec has been validated in table above)

On BTIP/BTIPI, only following configurations are allowed:

- G711A, G729
- G722, G711A
- G722, G711A, G729

On Btalk/BToI, only following configurations are allowed:

- G711 (A/μ), G729
- G722, G711 (A/μ)
- G722, G711 (A/μ), G729

3.4 Restrictions in Local Media Optimization

Some features may be unavailable due to Teams restrictions. Please check following link before applying LMO to your configuration:

<https://docs.microsoft.com/en-us/microsoftteams/direct-routing-media-optimization#known-issues>

3.5 Endpoints

As far as Local Media Optimization “Asia” model, that would allow direct RTP between Teams endpoint and BTIP/BTalk aSBC, is not certified, all Microsoft-certified endpoints are certified with Direct Routing and BTIP/BTalk. Indeed, RTP is at least screened by DR SBC.

4 AudioCodes SBC Configuration Checklist for BTIP/BTIPoI/BTalk/BToI

The checklist below presents all steps of configuration required for VISIT SIP Teams deployment.

4.1 Flow matrix with BToI

| Source | Source IP | Source port | Destination | Destination IP | Destination port | Comment |
|-------------|-----------------|------------------------------|-------------|-----------------|------------------------------|----------------|
| SBC | SBC public @IP | TLS 5061 | BToI | BToI public @IP | TLS 5061 | SIP Outgoing |
| BToI | BToI public @IP | Any | SBC | SBC public @IP | TLS 5061 | SIP Incoming |
| SBC | SBC public @IP | Range defined in Media Realm | BToI | BToI public @IP | UDP 6000-20000 | Media outgoing |
| BToI | BToI public @IP | UDP 6000-20000 | SBC | SBC public @IP | Range defined in Media Realm | Media incoming |

4.2 Flow matrix with BTIPoI

| Source | Source IP | Source port | Destination | Destination IP | Destination port | Comment |
|---------------|-------------------------|------------------------------|---------------|-------------------------|------------------------------|----------------|
| SBC | SBC public @IP | TLS 5061 | BTIPoI | BTIPoI SIP public @IP | TLS 5061 | SIP Outgoing |
| BTIPoI | BTIPoI SIP public @IP | Any | SBC | SBC public @IP | TLS 5061 | SIP Incoming |
| SBC | SBC public @IP | Range defined in Media Realm | BTIPoI | BTIPoI media public @IP | UDP 6000-38000 | Media outgoing |
| BTIPoI | BTIPoI media public @IP | UDP 6000-38000 | SBC | SBC public @IP | Range defined in Media Realm | Media incoming |

4.3 Configuration checklist for Office365 Tenant

Parameters to configure the SIP Trunk between Tenant and SBC:

| | |
|-----------------------|----------------------------|
| Fqdn | <Customer SBC Public FQDN> |
| SipSignallingPort | 5062 |
| MaxConcurrentSessions | <Number of max sessions> |
| Enabled | \$true |
| ForwardPai | \$true |
| SendSipOptions | \$true |
| ForwardCallHistory | \$true |



Powershell cmdlet:

```
#New-CsOnlinePSTNGateway -Fqdn sbc.contoso.com -SipSignallingPort 5062 -
MaxConcurrentSessions 50 -Enabled $true -ForwardPai $true -MediaBypass
$false -ForwardCallHistory $true -SendSipOptions $true
```

To activate media bypass:

```
#Set-CsOnlinePSTNGateway -Identity sbc.contoso.com -MediaBypass $true
```

Note: media bypass is only compatible from Audiocodes version 7.20A.254.475.

To activate local media optimization (Europe Scenario) at Tenant level:

```
#Set-CSOnlinePSTNGateway -Identity sbc.contoso.com -GatewaySiteID "MySite"
-ProxySBC $null -BypassMode Always
```

```
#New-CsTenantNetworkRegion -identity "MyRegion" -description "DR Media
Optimization SBCs for MyRegion"
```

```
#New-CsTenantNetworkSite -identity "MySite" -NetworkRegionID "MyRegion" -
description "DR MO Remote Site in MyRegion"
```

```
#New-CsTenantNetworkSubnet -identity <Network IP address> -MaskBits
<network range: example 24 or 25> -NetworksiteID "MySite" -description
"MySite Voice Subnet"
```

```
#New-CsTenantTrustedIPAddress -identity <Public NAT for MySite> -MaskBits
<network range: ex 24 or 25> -description "Public NAT for MySite"
```

Note: Local Media Optimization is only compatible from Audiocodes version 7.20A.258.475

4.4 Configuration checklist for QoS in Teams client

QoS management is done by configuring the Teams.exe at Windows level.

This configuration is done either locally or by GPO:

- Locally: Use policy-based Quality of Service (QoS) within Group Policy, and create a policy for Teams Audio with following parameters:

| Parameter | Value | Description |
|--------------------------|-------------|---|
| Policy Name | Teams Audio | |
| Application Name | Teams.exe | |
| Protocol | Both | TCP and UDP |
| Source Port Start | 50000 | Source ports used by Teams desktop clients are managed in the Teams Admin center. Microsoft recommends to keep this initial port range. |
| Source Port End | 50019 | |
| DSCP value | 46 | DSCP=46 Expedited Forwarding (EF) |



- By GPO: *#new-NetQosPolicy -Name "Teams Audio" -AppPathNameMatchCondition "Teams.exe" -IPProtocolMatchCondition Both -IPSrcPortStartMatchCondition 50000 -IPSrcPortEndMatchCondition 50019 -DSCPAction 46 -NetworkProfile All*

4.5 Parameters not available via the web admin page

Some parameters cannot be set via the Web Admin and must be configured via CLI or directly edited in the .INI file:

Parameter: `sbc-100trying-upon-reinvite`

Enables the device to send a SIP 100 Trying response upon receipt of a reINVITE request. BusinessTalk infrastructure must receive a response within the next 200ms following any INVITE/reINVITE. By default, the Audiocodes SBC does not generate 100 TRYING when receiving a reINVITE, enabling the parameter will force the SBC to generate it.

Via CLI: `configure voip > sbc settings > sbc-100trying-upon-reinvite on`

Via .ini file: inside the [SIP Params] section add the following line:

`SBC100TRYINGUPONREINVITE = 1`

4.6 SBC IP addressing requirements

For BT/BTIP:

- **Internet facing Interface:** 1 Public IP address + 1 Private IP address NATed to the Public IP
- Note: cf. NAT translation rule.
- **BT/BTIP SIP trunk Interface:** 1 IP address
- **Management Interface:** 1 IP address
- Note: For BT/BTIP customers, this interface is generally shared with the BT/BTIP SIP trunk interface. This assumption has been taken in this document.
- **[Optional] HA interface:** 1 IP address (non-routed)

For Business Talk over Internet (BTol), or BTIPol :

- **Internet facing Interface:** 1 Public IP
Notes:
 - For BTol/BTIPol customers, this interface is shared between TEAMS and BTol/BTIPol SIP Trunk.
 - BTol/BTIPol recommends not to use NAT (cf. NAT translation rule).
- **Management Interface:** 1 IP address
- **[Optional] HA interface:** 1 IP address (non-routed)



4.7 Configuration checklist for Mediant SBC – Standalone

Step 1 – IP Network configuration

Step 2 – Teams configuration

Step 3 – Business Talk configuration

Step 4 – Routing configuration

Step 5 – Pre-Recorded Tones files

| 4.7.1 Step 1 - IP Network configuration | |
|---|--|
| Ethernet Groups | |
| On the Mediant WebUi Interface: SETUP > IP Network > Core entities > Ethernet Groups | 1 Ethernet Group for Teams and BTol/BTIPol (GROUP_1) 1 Ethernet Group for BTIP or LAN (GROUP_2) [Optional] 1 Ethernet Group for HA (GROUP_3) <u>If BT/BTIP :</u> ✓ GROUP_1 is dedicated to Teams Only ✓ GROUP_2 is shared between BT/BTIP and Management <u>If BTol/BTIPol :</u> ✓ GROUP_1 is shared between Teams and BTol/BTIPol ✓ GROUP_2 is dedicated Management Only |
| Ethernet Devices | |
| SETUP > IP Network > Core entities > Ethernet Devices | 1 Ethernet Device for Teams and BTol/BTIPol (EthD_Teams) 1 Ethernet Device for BTIP or LAN (EthD_LAN or EthD_BTIP) [Optional] 1 Ethernet Device for HA (EthD_HA) <u>If BTIP :</u> ✓ EthD_BTIP associated to GROUP2 <u>If BTol/BTIPol :</u> ✓ EthD_Teams (shared with BTol): associated to GROUP1 ✓ EthD_LAN: associated to GROUP2 |
| IP Interfaces | |
| SETUP > IP Network > Core entities > IP Interface Devices | ✓ 1 IP Interface for Teams (IPInt_Teams) (shared with BTol/BTIPol) ✓ 1 IP Interface for LAN (IPInt_LAN or IPInt_BTIP) ✓ [Optional] 1 IP Interface for HA (IPInt_HA) ✓ IPInt_Teams: associated to “EthD_Teams”, Media and SIG <u>If BT/BTIP:</u> ✓ IPInt_BTIP: associated to “EthD_BTIP”, Media, SIG, OAM <u>If BTol / BTIPol:</u> ✓ IPInt_Teams: associated to “EthD_Teams”, Media, SIG |



| | |
|--|---|
| | <ul style="list-style-type: none"> ✓ IPInt_LAN: associated to "EthD_LAN", OAM <p><u>If HA mode :</u></p> <ul style="list-style-type: none"> ✓ IPInt_HA: associated to "EthD_HA", used for HA mode |
| NAT Translation | |
| <p>SETUP > IP Network > Core entities > NAT Translation</p> | <p><u>If BT/ BTIP:</u></p> <p>Create 1 rule for signalization and media traffic</p> <ul style="list-style-type: none"> - Source Interface : IPInt_Teams - Source Start Port : 1024 - Source End Port : 65535 - Target IP Address: <SBC Public IP Address> - Target Start Port: - Target End Port: <p><u>If BTol / BTIPol:</u></p> <p>General recommendation is not to use NAT translation for BTol/BTIPol. The SBC IP interface dedicated to Teams and BTol/BTIPol will be assigned the SBC Public IP address.</p> |
| Security | |
| TLS Contexts | |
| <p>On the Mediant WebUi Interface: SETUP > IP Network > Security > TLS Contexts</p> | <p>Create new TLS Context for Teams traffic (<i>shared with BTol/BTIPol</i>)</p> <ul style="list-style-type: none"> - Name: Teams-TLSContext - TLS Version: TLSv1.2 - DTLS Version: DTLSv1.2 - DH Key Size: 2048 |
| <p>SETUP > IP Network > Security > TLS Contexts > Teams-TLSContext > Change Certificate</p> | <p>Create a new CSR for Teams/BTol/BTIPol SBC FQDN and send it to the public certification authority for signing. Then upload it to the Mediant.</p> |
| <p>SETUP > IP Network > Security > TLS Contexts > Teams-TLSContext > Trusted Root Certificates</p> | <p>Import Root/ Intermediate Certificates</p> |
| Firewall [Warning: not fully tested] | |
| <p>On the Mediant WebUi Interface: SETUP > IP Network > Security > Firewall</p> | <p>Create new access lists for public DNS</p> <ul style="list-style-type: none"> - Index: 0 - Description: <DNS public IP> (eg. 8.8.8.8) - Source IP: <DNS public IP> (eg. 8.8.8.8) - Prefix Length: 32 - Use specific interface: Enable - Interface Name: IPInt_Teams - Action Upon Match: Allow <p>Create new access lists for each Teams proxy (6 in Total)</p> <ul style="list-style-type: none"> - Index: 1 to 6 - Description: <Teams proxy IP> (eg. 52.114.76.76) - Source IP: < Teams proxy IP > (eg. 52.114.76.76) - Prefix Length: 32 - Use specific interface: Enable - Interface Name: IPInt_Teams - Action Upon Match: Allow |



If BTol / BTIPol:

Create new access lists for each BTol/BTIPol aSBC (2 in total if nominal/backup)

- Index: **1**
- Description: **<BTOI / BTIPol public IP>**
- Source IP: **<BTOI / BTIPol public IP>**
- Prefix Length: **32**
- Use specific interface: **Enable**
- Interface Name: **IPInt_Teams**
- Action Upon Match: **Allow**

Create new access list to block all other networks (6 in Total). This rule has to be the last of the list.

- Index: **49**
- Description: **0.0.0.0**
- Source IP: **0.0.0.0**
- Prefix Length: **0**
- Use specific interface: **Enable**
- Interface Name: **IPInt_Teams**
- Action Upon Match: **Block**

4.7.2 Step 2 - Teams configuration

Allowed Audio Coder Groups

SETUP > Signaling&Media > Coders & Profiles > Allowed Audio Coders Groups

Create Teams_AudioCoders

Select the created entry, then click on "Allowed Audio Coders 0 Items" and click on new:

- Coder Name: **G722**
- Coder Name: **G711A-law**
- Coder Name: **G711U-law**
- Coder Name: **G729**

IP Profile

SETUP > Signaling&Media > Coders & Profiles > IP Profiles

Create new IP Profile for Teams

- Name: **Teams-IPProfile**
 - SBC Media Security Mode: **SRTP**
 - SBC Enforce MKI Size: **Enforce**
 - Reset SRTP Upon Re-key: **Disable**
 - Generate SRTP Keys Mode: **Only If Required**
 - Remote Early Media RTP Detection Mode: **By Media**
 - Allowed Audio Coder: **Teams_AudioCoders**
 - Allowed Coders mode: **Restriction and Preference**
 - RTCP Mode: **Generate Always**
 - ICE Mode: **Disable/Lite**
- (Lite only required when MediaBypass activated at Tenant level)
- RTP IP DiffServ: **46**
 - Signalling DiffServ: **24**
 - Remote Update Support: **Not Supported**
 - Remote re-INVITE: **Supported only with SDP**
 - Remote Delayed Offer Support: **Not Supported**
 - Remote REFER Mode: **Handle locally**
 - Remote Replaces Mode : **Handle Locally**
 - Remote 3xx Mode: **Handle locally**
 - Remote Hold Format: **Inactive**

If BTol/BTIPol, the following parameters must be changed:

- **Diversion Header Mode:** Remove
- **History-Info Header Mode:** Add

Media Realm

SETUP > Signaling&Media > Core Entities > Media Realms

Create new Media Realm for Teams traffic

- Name: **Teams-Media**
- Topology location: **UP**
- IPv4 Interface Name: **IPInt_Teams**
- UDP Port Range Start: **49160**
- Number of Media Session Legs: **1637**
- Default Media Realm: **Yes**

SIP Interface

SETUP > Signaling&Media > Core Entities > SIP Interface

Create new SIP Interface for Teams traffic

- Name : **Teams-SIPInterface**
- Topology Location: **UP**
- Network Interface : **IPInt_Teams**
- UDP Port : **0**

| | |
|---|--|
| | <ul style="list-style-type: none"> - TCP Port: 0 - TLS Port: 5062 - Enable TCP Keep alive: Enable - Classification Failure response Type: 0 - Media Realm: Teams-Media - TLS Context Name: Teams-TLSContext - TLS Mutual Authentication: Disable |
| Proxy Set | |
| SETUP > Signaling&Media > Core Entities > Proxy Set | Create new Proxy Set for Teams traffic <ul style="list-style-type: none"> - Name : Teams-Proxies - SBC IPv4 SIP Interface : Teams-SIPInterface - TLS Context Name: Teams-TLSContext - Proxy Keep-Alive : Using OPTIONS - Proxy Keep-Alive Time : 180 - Proxy Hot swap: Enable - Proxy Load Balancing method: Random Weights (Proxy Address Table) : Create 3 Entries for Teams Proxies <ul style="list-style-type: none"> - Index: 0 - Proxy Address: sip.pstnhub.microsoft.com:5061 - Transport Type: TLS - Priority Proxy: 1 - Proxy Random Weight: 1 - Index: 1 - Proxy Address: sip2.pstnhub.microsoft.com:5061 - Transport Type: TLS - Priority Proxy: 2 - Proxy Random Weight: 1 - Index: 2 - Proxy Address: sip3.pstnhub.microsoft.com:5061 - Transport Type: TLS - Priority Proxy: 3 - Proxy Random Weight: 1 - |
| Message Manipulation | |
| SETUP > Signaling&Media > Message Manipulations | Create new message manipulation " Teams_Privacy_Removal " <ul style="list-style-type: none"> - Name : Teams_Privacy_Removal - Manipulation Set ID: 1 - Row Role: Use Current Condition - Match> Message Type: Invite.Request - Match> Condition: Header.From.URL.user != 'anonymous' - Action> Action Subject: Header.Privacy - Action> Action Type: Remove - Action> Action Value: |
| SETUP > Signaling&Media > Message Manipulations | Create new message manipulation " M=x-data_Removal " <ul style="list-style-type: none"> - Name : M=x-data_Removal - Manipulation Set ID: 1 - Row Role: Use Current Condition - Match> Message Type: Reinvite |

| | |
|--|--|
| | <ul style="list-style-type: none"> - Match> Condition: Body.sdp regex (.*)(m=audio)(.)(m=x-data)(.)(.*) - Action> Action Subject: Body.sdp - Action> Action Type: Modify - Action> Action Value: \$1+\$2+\$3 |
| SETUP > Signaling&Media > Message Manipulations | <p>Create new message manipulation "Teams_Busy_Here_Cause34_Removal"</p> <ul style="list-style-type: none"> - Name: Teams_Busy_Here_Cause34_Removal - Manipulation Set ID: 1 - Row Role: Use Current Condition - Match> Message Type: Any - Match> Condition: Header.Request-URI.MethodType=='486' AND Header.Reason.Reason.Cause == '34' - Action> Action Subject: Header.Reason - Action> Action Type: Remove |
| SETUP > Signaling&Media > Message Manipulations | <p>Create new message manipulation "Keep_First_Codec"</p> <ul style="list-style-type: none"> - Name: Keep_First_Codec - Manipulation Set ID: 2 - Row Role: Use Current Condition - Match> Message Type: Any - Match> Condition: Body.sdp regex (.RTP/SAVP)([0-9]{1,3})(.)(101\r\n.*) - Action> Action Subject: Body.sdp - Action> Action Type: Modify - Action> Action Value: \$1+\$2+\$4 |
| SETUP > Signaling&Media > Message Manipulations | <p>Create new message manipulation "Add_Annexb_No_1"</p> <ul style="list-style-type: none"> - Name: Add_Annexb_No_1 - Manipulation Set ID: 1 - Row Role: Use Current Condition - Match> Message Type: Invite.Request - Match> Condition: Body.sdp regex (.)(m=audio)(.)(18)(.)(a=maxptime:200)(.)(.*) - Action> Action Subject: Body.sdp - Action> Action Type: Modify <p>Action> Action Value: \$0+'a=fmtp:18 annexb=no'+\$7</p> |
| SETUP > Signaling&Media > Message Manipulations | <p>Create new message manipulation "Add_Annexb_No_2"</p> <ul style="list-style-type: none"> - Name: Add_Annexb_No_2 - Manipulation Set ID: 1 - Row Role: Use Current Condition - Match> Message Type: Invite.Request - Match> Condition: Body.sdp regex (.)(a=fmtp:18 annexb=yes)(.)(.*) - Action> Action Subject: Body.sdp - Action> Action Type: Modify - Action> Action Value: \$1+'a=fmtp:18 annexb=no'+\$3 |
| IP Group | |
| SETUP > Signaling&Media > Core Entities > IP Group | <p>Create new IP Group for Teams traffic</p> <ul style="list-style-type: none"> - Name : Teams-IPGroup - Topology Location: UP - Proxy Set: Teams-Proxies - IP Profile: Teams-IPProfile - Media Realm: Teams-Media - SIP Group Name: <Customer Teams Public FQDN> |



| | |
|--|---|
| | <ul style="list-style-type: none"> - Classify by Proxy Set: Disable - Local Host Name: <Customer Teams Public FQDN> - Always use src Address: Yes - Media TLS Context: Teams-TLSContext - Proxy Keep-Alive using IP Group settings: Enable - Inbound Message Manipulation Set: 1 (Manipulation set id including manipulations:“Teams_Privacy_Removal”) - Outbound Message Manipulation Set: 2 (Manipulation set id including manipulations:“Teams_Privacy_Modify”) <p><u>If LMO (Europe Scenario):</u></p> <ul style="list-style-type: none"> - Internal Media Realm: BTIP-Media - Teams Media Optimization handling: Teams Decides - Teams Media Optimization Initial Behavior: Internal |
| Media Security | |
| SETUP > Signaling&Media > Media > IP Media Security | Configure following parameters: <ul style="list-style-type: none"> - Media Security : Enable |
| RTP / RTCP settings | |
| SETUP > Signaling&Media > Media > IP RTP/RTCP Settings | Configure following parameters: <ul style="list-style-type: none"> - RTP UDP Port Spacing : 10 - RFC2833 TX Payload Type: 101 - FC 2833 Rx Payload Type: 101 |
| DSP Settings | |
| SETUP > Signaling&Media > Media > DSP Settings | Configure following parameters: <ul style="list-style-type: none"> - Answer Detector Activity Delay: 512 - Answer Detector Silence Time: 96 - Answer Detector Sensitivity: 2 - Energy Detector Quality Factor: 0 - Energy Detector Threshold: 0 |
| Proxy & Registration | |
| SETUP > Signaling&Media > SIP Definitions > Proxy & Registration | Configure following parameters: <ul style="list-style-type: none"> - Gateway Name: <Customer SBC Teams Public FQDN> - Use Gateway Name for Options: Yes |
| Message Condition | |
| SETUP > Signaling&Media > SIP Definitions > Message Conditions | Create new message condition for incoming SIP messages <ul style="list-style-type: none"> - Name : Teams contact - Condition: header.contact.url.host contains 'pstnhub.microsoft.com' |
| Classification | |
| SETUP > Signaling&Media > SBC > Classification | Create new classification <ul style="list-style-type: none"> - Name : From Teams-IPGroup - Source SIP Interface: Teams-SIPInterface - Source IP Address: * - Destination Host: <Customer SBC Teams Public FQDN> - Message Condition: Teams contact - Source IP Group: Teams-IPGroup |
| SBC General Settings | |
| SETUP > Signaling&Media > SBC > SBC General Settings | Configure following parameters: <ul style="list-style-type: none"> - SBC Performance Profile: Optimized for SRTP |



4.7.3 Step 3 – Business Talk configuration

Allowed Audio Coder Groups

SETUP > Signaling&Media > Coders & Profiles > Allowed Audio Coders Groups

If BT:

Create BT_AudioCoders

Select the created entry, then click on “Allowed Audio Coders 0 Items” and click on new:

- Coder Name: **G722**
- Coder Name: **G711A-law**
- Coder Name: **G711U-law**
- Coder Name: **G729**

If BTIP:

Create BTIP_AudioCoders

Select the created entry, then click on “Allowed Audio Coders 0 Items” and click on new:

- Coder Name: **G722**
- Coder Name: **G711A-law**
- Coder Name: **G729**

If BTol/BTIPol:

Create BTol_AudioCoders (or BTIPol)

Select the created entry, then click on “Allowed Audio Coders 0 Items” and click on new:

- Coder Name: **G711A-law**

IP Profile

SETUP > Signaling&Media > Coders & Profiles > IP Profiles

Create new IP Profile for Business Talk

If BT:

- Name: **BTIP-IPProfile**
- SBC Media Security Mode: **RTP**
- Allowed Audio Coders: **AudioCoders_BT**

If BTIP:

- Name: **BTIP-IPProfile**
- SBC Media Security Mode: **RTP**
- Allowed Audio Coders: **AudioCoders_BTIP**

If BTol:

- Name: **BTol-IPProfile**
- SBC Media Security Mode: **SRTP**
- Allowed Audio Coders: **AudioCoders_BTol**

For Both BT/BTIP and BTol:

- Symmetric MKI: **Disable**
- MKI Size: **0**
- Allowed Coders Mode: **Restriction**
- RFC2833 DTMF Payload Type: **101**
- RTP IP DiffServ: **46**
- Signaling DiffServ: **24**
- P-Asserted-Identity header mode: **Add**



| | |
|---|--|
| | <ul style="list-style-type: none"> - Remote REFER Mode: Handle locally - Remote Replaces Mode : Handle Locally - Play RBT to Transferee : yes - Remote 3xx Mode: Handle locally - Remote Hold Format: Send Only |
| Media Realm | |
| <p>SETUP > Signaling&Media > Core Entities > Media Realms</p> | <p>Create new or edit Teams Media Realm for Business Talk</p> <p><u>If BT/BTIP :</u></p> <ul style="list-style-type: none"> - Name: BTIP-Media - Topology location: DOWN - IPv4 Interface Name: IPInt_BTIP - UDP Port Range Start: 16400 - Number of Media Session Legs: 1000 - Default Media Realm: No <p><u>If BTol /BTIPol (edit Teams):</u></p> <ul style="list-style-type: none"> - Name: Teams-Media - Topology location: UP - IPv4 Interface Name: IPInt_Teams - UDP Port Range Start: 6000 - Number of Media Session Legs: 5953 |
| SIP Interface | |
| <p>SETUP > Signaling&Media > Core Entities > SIP Interface</p> | <p>Create new SIP Interface for Business Talk</p> <p><u>If BT/BTIP:</u></p> <ul style="list-style-type: none"> - Name : BTIP-SIPInterface - Topology location: DOWN - Network Interface : IPInt_BTIP - TCP Port: 5060 - TLS Port: 0 - Media Realm: BTIP-Media <p><u>If BTol / BTIPol:</u></p> <ul style="list-style-type: none"> - Name : BTol-SIPInterface - Network Interface : IPInt_Teams - TCP Port: 0 - TLS Port: 5061 - Media Realm: Teams-Media - TLS Context Name: Teams-TLSContext - TLS Mutual Authentication: Enable <p><u>For Both BT/BTIP and BTol / BTIPol:</u></p> <ul style="list-style-type: none"> - Topology Location: DOWN - UDP Port : 0 |
| Proxy Set | |

| | |
|---|--|
| <p>SETUP > Signaling&Media > Core Entities > Proxy Set</p> | <p>Create new Proxy Set for Business Talk</p> <p><u>If BT/BTIP:</u></p> <ul style="list-style-type: none"> - Name : BTIP-Proxies - SBC IPv4 SIP Interface : BTIP-SIPInterface - Proxy Keep-Alive: Using OPTIONS - Proxy Keep-Alive Time: 300 - Redundancy Mode: Homing - Proxy Hot swap: Enable <p><u>If BTol:</u></p> <ul style="list-style-type: none"> - Name : BTol-Proxies - SBC IPv4 SIP Interface : BTol-SIPInterface - TLS Context Name : Teams-TLSContext - Proxy Keep-Alive : Using OPTIONS - Proxy Keep-Alive Time : 300 - Redundancy Mode : Homing - Proxy Hot swap : Enable <p><u>If BTIPol:</u></p> <ul style="list-style-type: none"> - Name : BTIPol-Proxies - SBC IPv4 SIP Interface : BTIPol-SIPInterface - TLS Context Name : Teams-TLSContext - Proxy Keep-Alive : Using OPTIONS - Proxy Keep-Alive Time : 300 - DNS Resolve Method : SRV <p>(Proxy Address Table) :</p> <p><u>If BT/BTIP:</u></p> <p>Create 2 Entries for Business Talk Proxies</p> <p>First, the Nominal entry (mandatory to get the higher priority)</p> <ul style="list-style-type: none"> - Proxy Address : <Nominal SBC ACME>:5060 - Transport Type : TCP <p>Then, the Backup entry (lower priority)</p> <ul style="list-style-type: none"> - Proxy Address : <Backup SBC ACME>:5060 - Transport Type : TCP <p><u>If BTol:</u></p> <p>Create the Entries for Business Talk Proxy</p> <ul style="list-style-type: none"> - Proxy Address : < Nominal BTol SBC FQDN >:5061 Transport Type: TLS - Proxy Address : < Backup BTol SBC FQDN >:5061 Transport Type: TLS <p><u>If BTIPol:</u></p> <p>Create the Entries for Business Talk Proxy</p> <ul style="list-style-type: none"> - Proxy Address : < BTIPol SBC FQDN SRV > - Transport Type: TLS |
| Message Manipulation | |
| <p>SETUP > Signaling&Media > Message Manipulations</p> | <p>Create new message manipulation "User-Agent_Modification »</p> <ul style="list-style-type: none"> - Name : User-Agent_Modification - Manipulation Set ID: 3 - Row Role: Use Current Condition |



| | |
|---|---|
| | <ul style="list-style-type: none"> - Match> Message Type: any - Match> Condition: - Action> Action Subject: Header.User-Agent - Action> Action Type: Modify - Action> Action Value: Header.User-Agent.Content + 'Teams' |
| SETUP > Signaling&Media > Message Manipulations | <p>Create new message manipulation "Add 101 to ACK SDP"</p> <ul style="list-style-type: none"> - Name: Add 101 to ACK SDP - Manipulation Set ID: 4 - Row Role: Use Current Condition - Match> Message Type: Ack - Match> Condition: Body.sdp !contains 'a=rtpmap:101 telephone-event/8000' - Action> Action Subject: Body.sdp - Action> Action Type: Add <p>Action> Action Value: 'a=rtpmap:101 telephone-event/8000\la=fmtp:101 0-15'</p> |
| SETUP > Signaling&Media > Message Manipulations | <p>For BTol/BTIPol only (not applicable to BTIP):</p> <p>Create new message manipulation "Add Diversion1 twds BTol"</p> <ul style="list-style-type: none"> - Name : Add Diversion1 twds BTol - Manipulation Set ID: 3 - Row Role: Use Current Condition - Match> Message Type: INVITE - Match> Condition: Header.History-Info exists - Action> Action Subject: Header.Diversion - Action> Action Type: Add <p>Action> Action Value: '<sip:temp@[SBC_Public_fqdn];user=phone>'</p> |
| SETUP > Signaling&Media > Message Manipulations | <p>For BTol/BTIPol only (not applicable to BTIP):</p> <p>Create new message manipulation "Modify Diversion1 twds BTol"</p> <ul style="list-style-type: none"> - Name : Modify Diversion1 twds BTol - Manipulation Set ID: 3 - Row Role: Use Current Condition - Match> Message Type: INVITE - Match> Condition: Header.History-Info.0 regex (<sip:)(.*)((@)(.)* - Action> Action Subject: Header.Diversion.URL.User - Action> Action Type: Modify - Action> Action Value: \$2 |
| SETUP > Signaling&Media > Message Manipulations | <p>For BTol/BTIPol only (not applicable to BTIP):</p> <p>Create new message manipulation "Remove History-Info twds BTol"</p> <ul style="list-style-type: none"> - Name : Remove History-Info twds BTol - Manipulation Set ID: 3 - Row Role: Use Current Condition - Match> Message Type: INVITE - Action> Action Subject: Header.History-Info - Action> Action Type: Remove |
| SETUP > Signaling&Media > Message Manipulations | <p>For BTol/BTIPol only (not applicable to BTIP):</p> <p>Create new message manipulation "Add Diversion2 twds BTol"</p> <ul style="list-style-type: none"> - Name : Add Diversion2 twds BTol - Manipulation Set ID: 3 |

| | |
|--|--|
| | <ul style="list-style-type: none"> - Row Role: Use Current Condition - Match> Message Type: INVITE - Match> Condition: Header.Referred-By exists - Action> Action Subject: Header.Diversion - Action> Action Type: Add Action> Action Value: '< sip:temp@[SBC_Public_fqdn];user=phone>' |
| SETUP > Signaling&Media > Message Manipulations | <p><u>For BTol/BTIPol only (not applicable to BTIP):</u> Create new message manipulation “Modify Diversion2 twds BTol»</p> <ul style="list-style-type: none"> - Name : Modify Diversion2 twds BTol - Manipulation Set ID: 3 - Row Role: Use Current Condition - Match> Message Type: INVITE - Match> Condition: Header.Referred-By exists - Action> Action Subject: Header.Diversion.URL.User - Action> Action Type: Modify - Action> Action Value: Header.Referred-By.URL.User |
| SETUP > Signaling&Media > Message Manipulations | <p><u>For BTol/BTIPol only (not applicable to BTIP):</u> Create new message manipulation “Remove Referred-By twds BTol»</p> <ul style="list-style-type: none"> - Name : Remove Referred-By twds BTol - Manipulation Set ID: 3 - Row Role: Use Current Condition - Match> Message Type: INVITE - Action> Action Subject: Header.Referred-By - Action> Action Type: Remove |
| IP Group | |
| SETUP > Signaling&Media > Core Entities > IP Group | Create new IP Group for Business Talk <p><u>If BT/BTIP:</u></p> <ul style="list-style-type: none"> - Name : BTIP-IPGroup - Proxy Set: BTIP-Proxies - IP Profile: BTIP-IPProfile - Media Realm: BTIP-Media - Media TLS Context: <empty> <p><u>If BTol / BTIPol:</u></p> <ul style="list-style-type: none"> - Name : BTol-IPGroup - Proxy Set: BTol-Proxies - IP Profile: BTol-IPProfile - Media Realm: Teams-Media - Media TLS Context: Teams-TLSContext <p><u>For Both BT/BTIP and BTol:</u></p> <ul style="list-style-type: none"> - Topology Location: DOWN - Outbound Message Manipulation Set: 3 (Manipulation set id including manipulations “User-Agent_Modification”) - Inbound Message Manipulation Set: 4 |

4.7.4 Step 4 - Routing configuration

IP to IP Routing



| | |
|---|---|
| <p>On the Mediant WebUi Interface: SETUP > Signaling&Media > SBC > Routing > IP-to-IP Routing</p> | <p>Create 4 IP to IP routing rules:</p> <p>1st one regarding OPTIONS Messages:</p> <ul style="list-style-type: none"> - Name: SIP-OPTIONS-Terminate - Match > Source IP Group: Any - Match > Request Type: OPTIONS - Match > ReRoute IP Group: Any - Action > Destination Type: Dest Address - Action > Destination Address: internal <p>2nd one regarding REFER Messages:</p> <ul style="list-style-type: none"> - Name: REFER-Terminate - Match > Source IP Group: Any - Match > Request Type: All - Match > Call Trigger: REFER - Match > ReRoute IP Group: Teams-IPGroup - Action > Destination Type: Request URI - Action > Destination IP Group: Teams-IPGroup <p>3rd one regarding <u>Business Talk</u> to <u>Teams</u> traffic:</p> <ul style="list-style-type: none"> - Name: BTIP(or BTol/BTIPol) to Teams - Match > Source IP Group: BTIP-IPGroup (or BTol-IPGroup) - Match > ReRoute IP Group: Any - Action > Destination IP Group: Teams-IPGroup <p>4th one regarding <u>Teams</u> to <u>Business Talk</u> traffic:</p> <ul style="list-style-type: none"> - Name: Teams to BTIP(or BTol/BTIPol) - Match > Source IP Group: Teams-IPGroup - Match > ReRoute IP Group: Any - Action > Destination IP Group: BTIP-IPGroup (or BTol-IPGroup) |
|---|---|

| | |
|--|---|
| 4.7.5 Step 5 – Pre-Recorded-Tones | |
| Auxiliary Files | |
| <p>On the Mediant WebUi Interface: SETUP>ADMINISTRATION>MAINTENANCE>Auxiliary Files</p> | <p>Load a Prerecorded Tones file in auxiliary files for ring back tone to be played on transfer scenarios (can be created with AudioCodes DConvert tool based on an audio file with the right codec, ex: G711)</p> |

4.8 Configuration checklist for Mediant SBC – HA

Step 1 – Configuration of the first SBC Mediant for HA

Step 2 – Configuration of the second SBC Mediant for HA

Step 3 – Initialize HA on the devices

4.8.1 Step 1 – Configuration of the first Mediant for HA

Note: During this stage, make sure that the second device is powered off or disconnected from network.

| Ethernet Groups | |
|---|---|
| On the Mediant WebUi Interface: SETUP > IP Network > Core entities > Ethernet Groups | Use a dedicated Ethernet Group for HA (GROUP_3) |
| Ethernet Devices | |
| SETUP > IP Network > Core entities > Ethernet Devices | Create 1 Ethernet Device for HA <ul style="list-style-type: none"> - Name: EthD_HA - VLAN ID: 3 - Underlying interface: GROUP3 - Tagging: Untagged - MTU: 1500 |
| IP Interfaces | |
| SETUP > IP Network > Core entities > IP Interface Devices | Create 1 IP Interface for HA (IPInt_HA) <ul style="list-style-type: none"> - Name: IPInt_HA - Application Type: MAINTENANCE - Ethernet Device: EthD_HA - IP Address: 192.168.0.1 (example) - Prefix length: 24 - Default Gateway: 0.0.0.0 |
| HA Settings | |
| SETUP > IP Network > Core entities > HA Settings | Configure following parameters: <ul style="list-style-type: none"> - HA Remote Address: 192.168.0.2 (example) - Preempt Mode: Disable / Enable (depends on customer's choice) - Preempt Priority: 5 (only if Preempt Mode is Enable) - HA Device Name: SBC1 - HA Remote Preempt Priority: 4 (only if Preempt Mode is Enable) - Redundant HA Device Name: SBC2 |

- Save the configuration to flash without RESET
- Power Down the first Mediant and move to next section (Step2)

4.8.2 Step 2 – Configuration of the second Mediant for HA

Note: During this stage, make sure that the first device is powered off or disconnected from network.

| Ethernet Groups | |
|-----------------|--|
|-----------------|--|



| | |
|---|---|
| On the Mediant WebUi Interface: SETUP > IP Network > Core entities > Ethernet Groups | Configuration must be identical comparing to the first Mediant. |
| Ethernet Devices | |
| SETUP > IP Network > Core entities > Ethernet Devices | Configuration must be identical comparing to the first Mediant. |
| IP Interfaces | |
| SETUP > IP Network > Core entities > IP Interface Devices | Create 1 IP Interface for HA (IPInt_HA) <ul style="list-style-type: none"> - Name: IPInt_HA - Application Type: MAINTENANCE - Ethernet Device: EthD_HA - IP Address: 192.168.0.2 (example) - Prefix length: 24 - Default Gateway: 0.0.0.0 |
| HA Settings | |
| SETUP > IP Network > Core entities > HA Settings | Configure following parameters: <ul style="list-style-type: none"> - HA Remote Address: 192.168.0.1 (example) - Preempt Mode: Disable / Enable (depends on customer's choice) - Preempt Priority: 4 (only if Preempt Mode is Enable) - HA Device Name: SBC2 - HA Remote Preempt Priority: 5 (only if Preempt Mode is Enable) - Redundant HA Device Name: SBC1 |

- Save the configuration to flash without RESET
- Power Down the second Mediant and move to next section (Step3)

4.8.3 Step 3 – Initialize HA on the devices

Note: You must connect both ports (two) in the Ethernet Group of the Maintenance interface to the network (i.e., two network cables are used). This provides 1+1 Maintenance port redundancy.

1. Cable the devices to the network.
2. Power up the devices; the redundant device synchronizes with the active device and updates its configuration according to the active device.
3. Access the active device with its' OAMP IP address and configure the device as required



4.9 . Configuration checklist for Mediant SBC – LMO Asia

4.9.1 Configuration for Proxy SBC

➤ Office365 Tenant

Tenant configuration for Proxy SBC remain the same as the one described in section [2.3](#)

Parameters to configure the SIP Trunk between Tenant and Downstream gateway:

| | |
|-----------------------|----------------------------------|
| Fqdn | <Downstream GW FQDN> |
| SipSignallingPort | 5062 |
| MaxConcurrentSessions | <Number of max sessions> |
| Enabled | \$true |
| ForwardPai | \$true |
| SendSipOptions | \$false |
| ForwardCallHistory | \$true |
| MediaBypass | \$true |
| GatewaySiteId | <SiteId> |
| ProxySBC | <Customer proxy SBC Public FQDN> |
| BypassMode | Always |

Powershell cmdlet:

```
# New-CsOnlinePSTNGateway -Identity <Downstream SBC FQDN> -SipSignalingPort 5062 -
ForwardCallHistory $true -ForwardPai $true -Enabled $true -MediaBypass $true -GatewaySiteId <SiteId>
-ProxySbc <Proxy SBC FQDN> -BypassMode Always -SendSipOptions $false
```

➤ Proxy SBC Teams side

Step 1 - Reuse the configuration checklist for Mediant SBC – Standalone section [2.6](#) except for IP Group and IP Profile entities

Step 2 - Modify the Teams IP Group and IP Profile for LMO Asia

IP Group

SETUP > Signaling&Media > Core Entities > IP Group

Create new IP Group for Teams traffic

- Name : **Teams-LMOAsia-IPGroup**
- Topology Location: **UP**
- Proxy Set: **Teams-Proxies**
- IP Profile: **Teams-IPProfile**
- Media Realm: **Teams-Media**
- Internal Media Realm: **BTIP-Media**

| | |
|---|---|
| | <ul style="list-style-type: none"> - Classify by Proxy Set: Disable - Local Host Name: <Customer Teams Public FQDN> - Always use src Address: Yes - Media TLS Context: Teams-TLSContext - Teams Local Media Optimization handling: Teams Decides - Teams Local Media Optimization Behavior: DirectMedia - Proxy Keep-Alive using IP Group settings: Enable - Inbound Message Manipulation Set: 1 (Manipulation set id including manipulations:“Teams_Privacy_Removal”) - Outbound Message Manipulation Set: 2 (Manipulation set id including manipulations:“Teams_Privacy_Modify”) |
| IP Profile | |
| SETUP > Signaling&Media > Coders & Profiles > IP Profiles | Create new IP Profile for Teams <ul style="list-style-type: none"> - Name: Teams-LMOAsia-IPProfile - SBC Media Security Mode: SRTP - SBC Enforce MKI Size: Enforce - Reset SRTP Upon Re-key: Disable - Generate SRTP Keys Mode: Only If Required - Remote Early Media RTP Detection Mode: By Media - Allowed Audio Coder: Teams_AudioCoders - Allowed Coders mode: Restriction and Preference - RTCP Mode: Generate Always - ICE Mode: Lite - RTP IP DiffServ: 46 - Signalling DiffServ: 24 - P-Asserted-Identity header mode: As is - SIP Update Support: Not Supported - Remote re-INVITE: Supported only with SDP - Remote Delayed Offer Support: Not Supported - Remote REFER Mode: Regular - Remote 3xx Mode: Transparent - Remote Hold Format: Inactive |

➤ **Proxy SBC Downstream GW side**

| | |
|---|---|
| DNS | |
| SETUP > IP Network > DNS > Internal DNS | Create new internal DNS row <ul style="list-style-type: none"> - Domain Name: <Downstream SBC FQDN> - First IP Address: <Downstream SBC IP> - Second IP Address: 0.0.0.0 - Third IP Address: 0.0.0.0 |
| Allowed Audio Coder Groups | |
| SETUP > Signaling&Media > Coders & Profiles > Allowed Audio Coders Groups | Create BT_AudioCoders or BTIP_AudioCoders Select the created entry, then click on “Allowed Audio Coders 0 Items” and click on new: If BTIP: <ul style="list-style-type: none"> - Coder Name: G722 - Coder Name: G711A-law - Coder Name: G711U-law - Coder Name: G729 If BT: |



| | |
|---|--|
| | <ul style="list-style-type: none"> - Coder Name: G722 - Coder Name: G711A-law - Coder Name: G729 |
| IP Profile | |
| SETUP > Signaling&Media > Coders & Profiles > IP Profiles | <p>Create new IP Profile for GW LMO Asia</p> <ul style="list-style-type: none"> - Name: GW-LMOAsia-IPProfile - SBC Media Security Mode: SRTP - SBC Enforce MKI Size: Don't Enforce - Generate SRTP Keys Mode: Only If Required - Allowed Audio Coder: BT_AudioCoders or BTIP_AudioCoders - Allowed Coders mode: Restriction and Preference - RFC2833 DTMF Payload Type: 101 - RTP IP DiffServ: 46 - Signalling DiffServ: 24 - P-Asserted-Identity header mode: Add - Play RBT to Transferee: yes - Remote REFER Mode: Regular - Remote 3xx Mode: Transparent - Remote Hold Format: Send Only |
| Media Realm | |
| SETUP > Signaling&Media > Core Entities > Media Realms | <p>Create new Media Realm for GW</p> <ul style="list-style-type: none"> - Name: GW-Media - Topology location: DOWN - IPv4 Interface Name: IPInt_GW - UDP Port Range Start: 16400 - Number of Media Session Legs: 1000 |
| SIP Interface | |
| SETUP > Signaling&Media > Core Entities > SIP Interface | <p>Create new SIP Interface for downstream GW traffic</p> <ul style="list-style-type: none"> - Name : GW-LMOAsia-SIPInterface - Topology Location: DOWN - Network Interface : IPInt_GW - UDP Port : 0 - TCP Port: 0 - TLS Port: 5061 - Enable TCP Keep alive: Enable - Classification Failure response Type: 500 - Media Realm: GW-Media - TLS Context Name: Teams-TLSContext - TLS Mutual Authentication: Disable |
| Proxy Set | |
| SETUP > Signaling&Media > Core Entities > Proxy Set | <p>Create new Proxy Set for GW traffic</p> <ul style="list-style-type: none"> - Name : GW-LMOAsia-Proxy - SBC IPv4 SIP Interface: GW-LMOAsia-SIPInterface - TLS Context Name: Teams-TLSContext - Proxy Keep-Alive: Using OPTIONS - Proxy Keep-Alive Time: 300 <p>(Proxy Address Table) : Create 1 Entry for GW Proxy</p> <ul style="list-style-type: none"> - Index: 0 - Proxy Address: <Gateway FQDN>:5061 - Transport Type: TLS |



| | |
|--|---|
| | <ul style="list-style-type: none"> - Priority Proxy: 1 - Proxy Random Weight: 1 |
| IP Group | |
| SETUP > Signaling&Media > Core Entities > IP Group | <p>Create new IP Group for GW traffic</p> <ul style="list-style-type: none"> - Name : GW-LMOAsia-IPGroup - Topology Location: DOWN - Proxy Set: GW-LMOAsia-Proxy - IP Profile: GW-LMOAsia-IPProfile - Media Realm: GW-Media - Outbound Message Manipulation Set: 3 - Internal Media Realm: GW-Media - Classify by Proxy Set: Enable - Media TLS Context: Teams-TLSContext |
| Message Condition | |
| SETUP > Signaling&Media > SIP Definitions > Message Conditions | <p>Create new message condition for incoming SIP messages</p> <ul style="list-style-type: none"> - Name : a teams contact - Condition: header.contact.url.host contains 'pstnhub.microsoft.com' |
| Classification | |
| SETUP > Signaling&Media > SBC > Classification | <p>Create new classification</p> <ul style="list-style-type: none"> - Name : From Teams-IPGroup - Source SIP Interface: Teams-SIPInterface - Source IP Address: * - Destination Host: <Customer Downstream private FQDN> - Message Condition: a teams contact <p>Source IP Group: Teams-IPGroup</p> |

4.9.2 Configuration for Downstream Gateway

Step 1 – General Settings

Step 2 – IP Network Configuration

Step 3a & b – Trunks Configuration: Proxy SBC side and BT/BTIP side

Step 4 – Routing configuration

| | |
|---|---|
| Step 1 – General Settings | |
| NTP Configuration | |
| On the Mediant WebUi Interface: SETUP > Administration > Time & Date | <ul style="list-style-type: none"> - Enable NTP: Enabled - Primary NTP Server Address: <IP Address> - Secondary NTP Server Address: <IP Address> - UTC Offset Time: Hours:0 Minutes:0 - Daylight Saving Time: Disable |
| Media Security | |
| SETUP > Signaling&Media > Media > Media Security | <ul style="list-style-type: none"> - Media Security: Enable |
| RTP/RTCP | |



| | |
|--|---|
| SETUP > Signaling&Media > Media > RTP/RTCP Settings | - RTP UDP Port Spacing: 10 |
| Proxy & Registration | |
| SETUP > Signaling&Media > SIP Definitions > Proxy&Registration | - Gateway Name: Gateway FQDN - Use Gateway Name for OPTIONS: Yes |

Step 2 – IP Network Configuration

Ethernet Groups

| | |
|---|--|
| On the Mediant WebUi Interface: SETUP > IP Network > Core entities > Ethernet Groups | ✓ 1 Ethernet Group shared between the Teams Proxy SBC and BT/BTIP (GROUP_1) |
|---|--|

Ethernet Devices

| | |
|---|---|
| SETUP > IP Network > Core entities > Ethernet Devices | ✓ 1 Ethernet Device shared between the Teams Proxy SBC and BT/BTIP (EthD) - Name: EthD - Underlying Interface: GROUP_1 - Tagging: Untagged - MTU: 1500 |
|---|---|

IP Interfaces

| | |
|---|--|
| SETUP > IP Network > Core entities > IP Interface Devices | ✓ 1 IP Interface shared between the Teams Proxy SBC and BT/BTIP () - Name: IPInt - Application Type: OAMP, Media, Control - Interface Mode: IPv4 manual - IP Address: <GW IP Address> - Prefix length: <Subnet Size> - Default Gateway: <GW default gateway> - Ethernet Device: EthD |
|---|--|

TLS Contexts

| | |
|--|--|
| On the Mediant WebUi Interface: SETUP > IP Network > Security > TLS Contexts | Create new TLS Context - Name: Self-Signed-TLS - TLS Version: TLSv1.2 - DTLS Version: DTLSv1.2 - DH Key Size: 2048 |
| SETUP > IP Network > Security > TLS Contexts > Teams-TLSContext > Change Certificate | Create a new CSR for Teams/BTol/BTIPol SBC FQDN and send it to the public certification authority for signing. Then upload it to the Mediant. |
| SETUP > IP Network > Security > TLS Contexts > Teams-TLSContext > Trusted Root Certificates | Import Root/ Intermediate Certificates |

Internal DNS

| | |
|--|--|
| On the Mediant WebUi Interface: SETUP > IP Network > DNS > Internal DNS | - Domain Name: <Proxy SBC FQDN> - First IP address: <Proxy SBC private IP> - Second IP Address: 0.0.0.0 - Third IP Address: 0.0.0.0 |
|--|--|

Step 3a - Trunks configuration – Proxy SBC side

SIP Interface



| | |
|---|---|
| <p>SETUP > Signaling&Media > Core entities > SIP Interface</p> | <p>Create new SIP Interface for the Teams Proxy SBC traffic</p> <ul style="list-style-type: none"> - Name: ProxyTeams-SIPInterface - Topology Location: UP - Network Interface: IPInt - UDP Port: 0 - TCP Port: 0 - TLS Port: 5061 - Enable TCP Keep alive: Enable - Classification Failure Response Type: 0 - Media Realm: GW-Media - TLS Context Name: Self-Signed-TLS - TLS Mutual Authentication: Disable |
| Proxy Set | |
| <p>SETUP > Signaling&Media > Core Entities > Proxy Set</p> | <p>Create new Proxy Set for the Teams proxy traffic</p> <ul style="list-style-type: none"> - Name: ProxyTeams-Proxy - SBC IPv4 SIP Interface: ProxyTeams-SIPInterface - TLS Context Name: Self-Signed-TLS - Proxy Keep-Alive: Using OPTIONS - Proxy Keep-Alive Time: 300 - Proxy Hot swap: Enable <p>(Proxy Address Table): Create 1 Entry for the Teams Proxy SBC</p> <ul style="list-style-type: none"> - Index: 0 - Proxy Address: <Proxy SBC FQDN @IP>5061 - Transport Type: TLS - Priority Proxy: 1 - Proxy Random Weight: 1 |
| Media Realms | |
| <p>SETUP > Signaling&Media > Core Entities > Media Realms</p> | <p>Create new Media Realm for the Teams proxy traffic</p> <ul style="list-style-type: none"> - Name: GW-Media - Topology location: UP - IPv4 Interface Name: IPInt - UDP Port Range Start: 16400 - Number of Media Session Legs: 1000 |
| IP Group | |
| <p>SETUP > Signaling&Media > Core Entities > IP Group</p> | <p>Create new IP Group for the Teams proxy traffic</p> <ul style="list-style-type: none"> - Name : ProxyTeams-IPG - Topology Location: UP - Proxy Set: ProxyTeams-Proxy - IP Profile: ProxyTeams-IPProfile - Media Realm: GW-Media - Internal Media Realm: GW-Media - SIP Group Name: <Downstream SBC FQDN> - Classify by Proxy Set: Enable - Local Host Name: <Downstream SBC FQDN> - Always use src Address: Yes - Media TLS Context: Self-Signed-TLS - Teams Local Media Optimization Initial Behavior: DirectMedia - Proxy Keep-Alive using IP Group settings: Enable |
| IP Profile | |

| | |
|---|---|
| <p>SETUP > Signaling&Media > Coders & Profiles > IP Profiles</p> | <p>Create new IP Profile for the Teams proxy traffic</p> <ul style="list-style-type: none"> - Name: ProxyTeams-IPProfile - SBC Media Security Mode: Secured - Remote Early Media RTP Detection Mode: By Media - Allowed Audio Coder: Teams_AudioCoders - Allowed Coders mode: Restriction and Preference - RTCP Mode: Generate Always - ICE Mode: Lite - SIP Update Support: Not Supported - Remote re-INVITE: Supported only with SDP - Remote Delayed Offer Support: Not Supported - Remote Representation Mode: Replace contact - Remote Replaces Mode: Handle locally - Remote REFER Mode: Handle locally - Remote 3xx Mode: Handle locally - Play RBT To Transferee: No - Remote Hold Format: Inactive |
| Coders | |
| <p>SETUP > Signaling&Media > Coders & Profiles > Allowed Audio Coders Groups</p> | <p>Create new allowed coders for the Teams proxy traffic</p> <ul style="list-style-type: none"> - Name: Teams-Audiocodes - Allowed Audio Coder Item: G722, G711A-Law, G711U-Law, G729 |

Step 3b - Trunks configuration – BT/BTIP side

| | |
|---|---|
| SIP Interface | |
| <p>SETUP > Signaling&Media > Core entities > SIP Interface</p> | <p>Create new SIP Interface for BT/BTIP</p> <ul style="list-style-type: none"> - Name: BTIP-SIPInterface - Topology Location: UP - Network Interface: IPInt - UDP Port: 0 - TCP Port: 5060 - TLS Port: 0 - Enable TCP Keep alive: Enable - Classification Failure Response Type: 0 - Media Realm: GW-Media |
| Proxy Set | |
| <p>SETUP > Signaling&Media > Core Entities > Proxy Set</p> | <p>Create new Proxy Set for BT/BTIP</p> <ul style="list-style-type: none"> - Name: BTIP-Proxy - SBC IPv4 SIP Interface: ProxyTeams-SIPInterface - TLS Context Name: Self-Signed-TLS - Proxy Keep-Alive: Using OPTIONS - Proxy Keep-Alive Time: 300 - Redundancy Mode: Homing - Proxy Hot swap: Enable <p>(Proxy Address Table): Create 2 Entries for nominal/backup BT/BTIP infra</p> <ul style="list-style-type: none"> - Index: 0 - Proxy Address: <Nominal SBC ACME>:5060 - Transport Type: TCP - Priority Proxy: 1 - Proxy Random Weight: 1 |



| | |
|---|---|
| | <ul style="list-style-type: none"> - Index: 1 - Proxy Address: <Backup SBC ACME>:5060 - Transport Type: TCP - Priority Proxy: 2 - Proxy Random Weight: 1 |
| IP Group | |
| SETUP > Signaling&Media > Core Entities > IP Group | <p>Create new IP Group for BT/BTIP</p> <ul style="list-style-type: none"> - Name: BTIP-IPG - Topology Location: UP - Proxy Set: BTIP-Proxy - IP Profile: BTIP-IPProfile - Media Realm: GW-Media - Outbound Message Manipulation Set: 1 - Internal Media Realm: GW-Media - Classify by Proxy Set: Enable |
| IP Profile | |
| SETUP > Signaling&Media > Coders & Profiles > IP Profiles | <p>Create new IP Profile for BT/BTIP</p> <ul style="list-style-type: none"> - Name: ProxyTeams-IPProfile - SBC Media Security Mode: Secured - Remote Early Media RTP Detection Mode: By Media - Allowed Audio Coder: Teams_AudioCoders - Allowed Coders mode: Restriction and Preference - RTCP Mode: Generate Always - ICE Mode: Lite - SIP Update Support: Not Supported - Remote re-INVITE: Supported only with SDP - Remote Delayed Offer Support: Not Supported - Remote Representation Mode: Replace contact - Remote Replaces Mode: Handle locally - Remote REFER Mode: Handle locally - Remote 3xx Mode: Handle locally - Play RBT To Transferee: No - Remote Hold Format: Inactive |
| Coders | |
| SETUP > Signaling&Media > Coders & Profiles > Allowed Audio Coders Groups | <p>Create new allowed coders for BTIP</p> <ul style="list-style-type: none"> - Name: BTIP-Audiocodes - Allowed Audio Coder Item: G722, G711A-Law, G729 <p>Or</p> <p>Create new allowed coders for BT</p> <ul style="list-style-type: none"> - Name: BT-Audiocodes <p>Allowed Audio Coder Item: G722, G711A-Law, G711U-Law, G729</p> |
| Message Manipulation | |
| SETUP > Signaling&Media > Message Manipulation > Message Manipulations | <p>Create new message manipulation</p> <ul style="list-style-type: none"> - Name: User-Agent_Modification - Manipulation Set ID: 1 - Message Type: Any - Action Subject: Header.User-Agent - Action Type: Modify - Action Value: Header.User-Agent.Content + ' Downstream' |

Step 4 - Routing configuration

IP to IP Routing

On the Mediant WebUi Interface:
SETUP > Signaling&Media > SBC >
Routing > IP-to-IP Routing

Create 3 IP to IP routing rules:

First one regarding REFER Messages:

- Name: **REFER-Terminate**
- Match > Source IP Group: **Any**
- Match > Request Type: **All**
- Match > Call Trigger: **REFER**
- Match > ReRoute IP Group: **ProxyTeams-IPG**
- Action > Destination Type: **IPGroup**
- Action > Destination IP Group: **ProxyTeams-IPG**

2nd one regarding BT/BTIP to Proxy Teams traffic:

- Name: **From BTIP to Proxy Teams**
- Match > Source IP Group: **BTIP-IPG**
- Match > ReRoute IP Group: **Any**
- Action > Destination IP Group: **ProxyTeams-IPG**

3rd one regarding Proxy Teams to BT/BTIP traffic:

- Name: **From Proxy Teams to BTIP**
- Match > Source IP Group: **ProxyTeams-IPG**
- Match > ReRoute IP Group: **Any**
- Action > Destination IP Group: **BTIP-IPG**



4.10 AudioCodes Analog Phones configuration checklist

4.10.1 Architecture “Type 1” - Analog Phones connected to DR SBC through remote MediaPack (ATA)

Checklist:

Step 1 – Setup DR SBC trunk configuration towards ATA

Step 2 – Configure ATA to connect with DR SBC gateway

SIP-trunk configuration From DR SBC to MediaPack gateway

| Step 1 - Setup DR SBC trunk configuration towards ATA | |
|--|--|
| IP Interface | |
| On the vSBC DR WebUi Interface: SETUP > IP NETWORK > CORE ENTITIES > IP interface | 1 IP Interface for ATA (MPxxx-IP_Inter) <ul style="list-style-type: none"> - Name: MPxxx-IP_Inter - Application Type: Media + Control - Ethernet Device: <Eth_dev to MPxxx> - Interface Mode: IPv4 Manual - IP Address: DR_SBC @IP - Prefix Length: <Mask> - Default Gateway: <Net. GW> |
| Media Realm | |
| SETUP > SIGNALING&MEDIA > CORE ENTITIES > Media Realms | Create new Media Realm for ATA (BT media profile can be re-use) <ul style="list-style-type: none"> - Name: BT Media - IPv4 Interface Name: MPxxx-IP_Inter - UDP Port Range Start: 6000 - Number of Media Session Legs: 1000 - UDP Port Range End: 6999 - TCP Port Range Start: 0 - TCP Port Range End: 0 |
| SIP Interfaces | |
| SETUP > SIGNALING&MEDIA > CORE ENTITIES > SIP Interface | Create SIP Interface for ATA traffic <ul style="list-style-type: none"> - Name : MPxxx-SIP_Inter - Network Interface : MPxxx-IP_Inter - Application Type: SBC - UDP Port : 5060 - TCP Port: 5060 - Media Realm: BT Media - Direct Media: Enable |
| Proxy Set & Proxy IP | |
| SETUP > SIGNALING&MEDIA > CORE ENTITIES > Proxy Sets | Create Proxy Set for ATA traffic <ul style="list-style-type: none"> - Name : MPxxx-ProxySet - SBC IPv4 SIP Interface : MPxxx-SIP_Inter - Proxy Address/Port: <MP@IP>:5060 - Transport Type: TCP |



| IP Group | |
|---|---|
| SETUP > SIGNALING&MEDIA > CORE ENTITIES > IP Groups | Create IP Group for ATA <ul style="list-style-type: none"> - Name: MPxxx-IPG - Type: Server: Server - Proxy Set: MPxxx-ProxySet - IP Profile: <GW IP Profile> - Media Realm: BT Media |
| IP to IP Routing | |
| | Create IP to IP Routing from Teams to ATA <ul style="list-style-type: none"> - Name: Teams-to-MPxxx - Source IP Group: Any - Destination Username Pattern: <Pattern> - Destination Type: IP Group - Destination IP Group: MPxxx-IPG |

SIP-trunk configuration From MediaPack to DR SBC gateway

| Step 2 - Configure ATA to connect with DR SBC gateway | |
|---|---|
| General Parameters | |
| On the MediaPack WebUi Interface: CONFIGURATION > SIP DEFINITIONS > General Parameters | Configure General Parameters on MediaPack <ul style="list-style-type: none"> - Enable Early Media: Enable - SIP TCP Local Port: 5060 - SIP TCP Destination Port: 5060 |
| Coders Profile | |
| CONFIGURATION > CODERS and PROFILES > Coders | Configure Coders for MediaPack <ul style="list-style-type: none"> - 1st Coder Name: G.711A-law - Packetization Time: 20 - Rate: 64 - Payload Type: 8 - Silence Suppression: Disabled - 2nd Coder Name: G.729 - Packetization Time: 20 - Rate: 8 - Payload Type: 18 - Silence Suppression: Disabled |
| Coders Group Setting | |
| CONFIGURATION > CODERS and PROFILES > Coders Group Settings | ✓ Configure Coders Group for MediaPack <ul style="list-style-type: none"> - Coder Group ID: 1 - 1st coder name: G.711A-law - Packetization Time: 20 - Rate: 64 - Payload Type: 8 - Silence Suppression: Disabled - 2nd coder: G.729 - Packetization Time: 20 - Rate: 8 - Payload Type: 18 - Silence Suppression: Disabled |
| Tel Profile | |



| | |
|---|---|
| CONFIGURATION > CODERS and PROFILES > Tel Profile Settings | <ul style="list-style-type: none"> ✓ Create Tel Profile on MediaPack - Profile ID: <1,2,3...> - Profile Name: TelIP_AD - Enable Early Media: Enable - Coder Group: <1,2,3...> |
| IP Profile | |
| CONFIGURATION > CODERS and PROFILES > IP Profile Settings | <ul style="list-style-type: none"> ✓ Create IP Profile on MediaPack - Profile ID: <1,2,3...> - Profile Name: IPP_AD - Early Name: Enable - Coder Group: <1,2,3...> |
| Hunt Group Routing | |
| CONFIGURATION > GW and IP to IP > Routing > IP to Hunt Group Routing | <ul style="list-style-type: none"> ✓ Create Hunt Group on MediaPack - Dest. Phone Prefix: <phone prefix> - Src. Phone Prefix: * - Src. IP Address: * - Hunt Group: <1,2,3...> |
| Endpoint Phone Number | |
| CONFIGURATION > GW and IP to IP > HUNT GROUP > Endpoint Phone Number | <ul style="list-style-type: none"> ✓ Configure Analog Phone Number on MediaPack - Channel(s) : <1,2,3...> - Phone Number: <number> - Hunt Group: <1,2,3...> - Tel Profile ID: <1,2,3...> |
| Number Manipulation for IP->Tel | |
| CONFIGURATION > GW and IP to IP > Manipulations > Dest Number IP->Tel | <ul style="list-style-type: none"> ✓ Create number manipulation from IP to Tel - Index : <1,2,3...> - Dest Prefix: <1,2,3...> - Stripped Digits from Left: <...> - Prefix to add: <...> |
| CONFIGURATION > GW and IP to IP > Manipulations > Dest Number Tel->IP | <ul style="list-style-type: none"> ✓ Create number manipulation from Tel to IP - Index : <1,2,3...> - Dest Prefix: <1,2,3...> - Stripped Digits from Left: <...> - Prefix to add: <...> |
| Routing | |
| CONFIGURATION > GW and IP to IP > Routing > Tel to IP Routing | <ul style="list-style-type: none"> ✓ Create call routing between MediaPack and DR SBC - Src. Hunt Group: <1,2,3...> - Dest. Phone Prefix: <Prefix> - Src. Phone Prefix: * - Dest. IP Address: <IP DR SBC> - Port: 5060 - Transport Type: UDP - IP Profile ID: <1,2,3...> - Media Realm: Teams-Media |
| DTMF | |
| CONFIGURATION > GW and IP to IP > DTMF and Supplementary > DTMF & Dialing | <ul style="list-style-type: none"> ✓ Configure DTMF on Media Pack - 1st Tx DTMF Option: RFC 2833 - RFC 2833 Payload Type: 101 |



4.10.2 Architecture “Type 2” - Analog Phones connected to DR SBC through remote Local gateway (e.g. AudioCodes Mediant)

Checklist:

- Step 1 – Setup DR SBC trunk configuration towards Local gateway
- Step 2 – Configure AudioCodes Local Gateway to connect with DR SBC gateway
- Step 3 – Configure FXS Interface on Local gateway

SIP-trunk configuration From DR SBC to Local gateway

| Step 1 - Setup DR SBC trunk configuration towards AudioCodes Local Gateway | |
|--|--|
| IP Interface | |
| On the vSBC DR WebUi Interface: SETUP > IP NETWORK > CORE ENTITIES > IP interface | 1 IP Interface for Local GW (LocalGW-IP_Inter) <ul style="list-style-type: none"> - Name: LocalGW-IP_Inter - Application Type: Media + Control - Ethernet Device: <Eth_dev to Local GW> - Interface Mode: IPv4 Manual - IP Address: DR_SBC @IP - Prefix Length: <Mask> - Default Gateway: <Net. GW> |
| Media Realm | |
| SETUP > SIGNALING&MEDIA > CORE ENTITIES > Media Realms | Create new Media Realm for Local GW (BT media profile can be re-use) <ul style="list-style-type: none"> - Name: GWAC-Media - IPv4 Interface Name: LocalGW-IP_Inter - UDP Port Range Start: 6000 - Number of Media Session Legs: 1000 - UDP Port Range End: 6999 - TCP Port Range Start: 6000 - TCP Port Range End: 6999 |
| SIP Interfaces | |
| SETUP > SIGNALING&MEDIA > CORE ENTITIES > SIP Interface | Create SIP Interface for Local GW traffic <ul style="list-style-type: none"> - Name : LocalGW-SIP_Inter - Network Interface : LocalGW-IP_Inter - Application Type: SBC - UDP Port : 5060 - TCP Port: 5060 - Media Realm: BT Media - Direct Media: Enable |
| Proxy Set & Proxy IP | |
| SETUP > SIGNALING&MEDIA > CORE ENTITIES > Proxy Sets | Create Proxy Set for Local GW traffic <ul style="list-style-type: none"> - Name : LocalGW-ProxySet - SBC IPv4 SIP Interface : LocalGW-IP_Inter - Proxy Address/Port: <MP@IP>:5060 |



| | |
|---|---|
| | - Transport Type: TCP |
| IP Group | |
| SETUP > SIGNALING&MEDIA > CORE ENTITIES > IP Groups | <p>Create IP Group for Local GW</p> <ul style="list-style-type: none"> - Name: LocalGW-IPG - Type: Server: Server - Proxy Set: LocalGW-ProxySet - IP Profile: <GW IP Profile> - Media Realm: GWAC-Media |
| IP to IP Routing | |
| | <p>Create IP to IP Routing from Teams to Local GW</p> <ul style="list-style-type: none"> - Name: Teams-to-LocalGW - Source IP Group: Any - Destination Username Pattern: <Pattern> - Destination Type: IP Group - Destination IP Group: LocalGW-IPG |

SIP-trunk configuration From Local gateway to DR SBC gateway

Step 2 - Configure AudioCodes Local Gateway to connect with DR SBC gateway

| | |
|--|---|
| IP Interface | |
| On the Local gateway WebUi Interface: SETUP > IP NETWORK > CORE ENTITIES > IP interface | <p>1 IP Interface on Local GW (GW-IP_Inter)</p> <ul style="list-style-type: none"> - Name: GW-IP_Inter - Application Type: Media + Control - Ethernet Device: <Eth_dev to Local GW> - Interface Mode: IPv4 Manual - IP Address: LocalGW_SBC @IP - Prefix Length: <Mask> - Default Gateway: <Net. GW> |
| Media Realm | |
| SETUP > SIGNALING&MEDIA > CORE ENTITIES > Media Realms | <p>Create new Media Realm on Local GW (BT media profile can be re-use)</p> <ul style="list-style-type: none"> - Name: DRSBC-Media - IPv4 Interface Name: GW-IP_Inter - Port Range Start: 6000 - Number of Media Session Legs: 1000 |
| SIP Interfaces | |
| SETUP > SIGNALING&MEDIA > CORE ENTITIES > SIP Interface | <p>Create SIP Interface on Local GW traffic</p> <ul style="list-style-type: none"> - Name : DRSBC-SIP_Inter - Network Interface : GW-IP_Inter - Application Type: GW - UDP Port : 5060 - TCP Port: 5060 - Media Realm: DRSBC-Media - Direct Media: Disable |
| Proxy Set & Proxy IP | |
| SETUP > SIGNALING&MEDIA > CORE ENTITIES > Proxy Sets | <p>Create Proxy Set on Local GW traffic</p> <ul style="list-style-type: none"> - Name : DR_SBC-ProxySet - SBC IPv4 SIP Interface : DRSBC-SIP_Inter - Proxy Address/Port: <DR_SBC@IP>:5060 - Transport Type: TCP |

| IP Group | |
|---|--|
| SETUP > SIGNALING&MEDIA > CORE ENTITIES > IP Groups | Create IP Group on Local GW <ul style="list-style-type: none"> - Name: DR_SBC-IPG - Type: Server: Server - Proxy Set: DR_SBC-ProxySet - IP Profile: <GW IP Profile> - Media Realm: DRSBC-Media |

FXS interfaces on Local gateway

| Step 3 - FXS ports on Local gateway | |
|---|--|
| Tel Profiles | |
| On the Local gateway WebUi Interface: SETUP > SIGNALING&MEDIA > CODERS&PROFILES > Tel Profiles | Configure Tel Profile on Local GW <ul style="list-style-type: none"> - Name: AD-TelProfile - CodersGroup: <GW_coders_group> - Enable Early Media: Enable |
| Trunk Groups Settings | |
| SETUP > SIGNALING&MEDIA > GATEWAY > Trunk Groups Settings | Create new Trunk Group on Local gateway <ul style="list-style-type: none"> - Name: AD - Trunk Group ID: <1,2,3...> - Channel Select Mode: By Dest Phone Number - Registration Mode: Don't Register |
| Trunk Groups | |
| SETUP > SIGNALING&MEDIA > GATEWAY > Trunk Groups | Create SIP Interface on Local GW traffic <ul style="list-style-type: none"> - Module : Module FXS - Channels: <1,2,3...> - Phone Number: <xxxxxxxx> - Trunk Group ID: <ID> - Tel Profile Name: AD-TelProfile |

4.10.3 Architecture “Type 3” Analog Phone connected to DR SBC - AudioCodes hardware (SBC+MGW)

 **This architecture inferred from previous one, but actually not tested in lab.**

Checklist:

Step 1 – Setup DR SBC hardware (SBC+MGW) as “Mediant SBC – Standalone” – follow the checklist showing in 2.2 chapter from [VISIT SIP Teams – Configuration Checklist](#) [4] document:

Step 1.1 – IP Network configuration

Step 1.2 – Teams configuration

Step 1.3 – Business Talk configuration

Step 1.4 – Routing configuration



Step 1.5 – Pre-Recorded Tones files

Step 2 – Configure FXS Interface on DR SBC hardware – follow the checklist from [2.2.3](#) chapter of this document.

4.10.4 Configure Flash Hook

Configure Flash Hook on the AudioCodes local Gateway

Step 1 - FXS ports on Local gateway

Tel Profiles

On the Local gateway WebUi Interface:
SETUP >

Configure FlasHook on Local GW

- Enable Call Transfer Using Reinvites: **1**
- Flash Keys Sequence Style: **1**
- Flash Keys SequenceTimeout: **2000**

Configure Flash Hook on the MediaPack (ATA)

 **The Flash Hook features does not work for Analog Phones connected to MediaPack gateway.**

4.11 AudioCodes FAX configuration checklist

4.11.1 FXS fax on Mediant configuration

➤ Telephony profile

The FXS ports with fax devices connected requires dedicated configuration for fax. To create TelProfile go to **SETUP > SIGNALING & MEDIA > CODERS & PROFILES > Tel Profiles**.

Create new profile by pressing **+ New** and set:

| Parameter | Value | Description |
|----------------------|-------------------|---|
| Name | TelProfile_FXSFAX | Profile name |
| Fax Signaling Method | T.38 Relay | Select T.38 protocol for fax transmission |

➤ FXS port configuration update

Go to **SETUP > SIGNALING & MEDIA > GATEWAY > Trunks & Groups > Trunk Groups**

Update TEL PROFILE NAME on chosen trunk group to **TelProfile_FXSFAX**

➤ Update IP Profile

Note

[Please note that there are differences for BT and BTIP configuration for this point.](#)

Configuration for BT architecture

Go to **SETUP > SIGNALING & MEDIA > CODERS & PROFILES > IP Profiles**.

Select profile defined for Business Talk IP Group and update parameters:

| Parameter | Value | Description |
|-------------------------------------|------------|---|
| MEDIA SECURITY | | |
| SBC Media Security Mode | RTP | Disable secured RTP to avoid TLS in SDP |
| Gateway Media Security Mode | Disable | Disable secured RTP to avoid TLS in SDP |
| SBC FAX | | |
| Remote Renegotiate on fax detection | No | Describes if the remote renegotiate on fax detection |
| GATEWAY FAX AND MODEM | | |
| Fax Signaling Method | T.38 Relay | Use T38 for fax transmission |

Configuration for BTIP architecture

Go to **SETUP > SIGNALING & MEDIA > CODERS & PROFILES > IP Profiles**.

Select profile defined for Business Talk IP Group and update parameters:

| Parameter | Value | Description |
|-----------------------------|-------------------|---|
| MEDIA SECURITY | | |
| SBC Media Security Mode | RTP | Disable secured RTP to avoid TLS in SDP |
| Gateway Media Security Mode | Disable | Disable secured RTP to avoid TLS in SDP |
| GATEWAY FAX AND MODEM | | |
| Fax Signaling Method | T.38 Relay | Use T38 for fax transmission |

➤ General fax parameters

Note

[Please note that there are differences for BT and BTIP configuration for this point.](#)

Configuration for BT architecture

Go to **SETUP > SIGNALING & MEDIA > MEDIA > Fax/Modem/CID Settings** and update:

| Parameter | Value | Description |
|-------------------------------------|-------------------|--|
| Fax Transport Mode | T.38 Relay | Use T38 for fax transmission |
| CNG Detector Mode | Event only | Determines the fax CNG tone detector mode. |
| Fax Relay Redundancy Depth | 1 | Set pages transmission redundancy |
| Fax Relay Enhanced Redundancy Depth | 4 | Set fax negotiation redundancy |
| Fax/Modem Bypass Coder Type | 8 | Sets the Fax/Modem bypass coder |

Go to **SETUP > SIGNALING & MEDIA > MEDIA > RTP/RTCP Settings** and update:

| Parameter | Value | Description |
|---------------------------|----------|---|
| Modem Bypass Payload Type | 8 | Modem Bypass (VBD) Payload type. |

The next, EnableFaxModemInbandNetworkDetection parameter can be set only using CLI/configuration file and is not visible in web application. To set this parameter go to dedicated configuration page: <https://<MediantIP>/AdminPage> (note: subpage address is case sensitive).

Go to “ini Parameters” subsite using left sided menu.

Parameter name: **EnableFaxModemInbandNetworkDetection**

Enter value: **1**

Click “Apply New Value”.

If parameter is set correctly you should see output:

```
Parameter Name: ENABLEFAXMODEMINBANDNETWORKDETECTION
Parameter New Value: 1
Parameter Description: Enables or disables inband network detection related to fax/modem.
```

Configuration for BTIP architecture

Go to **SETUP > SIGNALING & MEDIA > MEDIA > Fax/Modem/CID Settings** and update:

| Parameter | Value | Description |
|-------------------------------------|------------|-----------------------------------|
| Fax Transport Mode | T.38 Relay | Use T38 for fax transmission |
| Fax Relay Redundancy Depth | 1 | Set pages transmission redundancy |
| Fax Relay Enhanced Redundancy Depth | 4 | Set fax negotiation redundancy |

➤ **Routing**

The routing of fax calls must be reconfigured to bypass Mediation Server. Go to **SETUP > SIGNALING & MEDIA > GATEWAY > Routing > Tel->IP Routing**. Select line assigned to chosen FXS or create new one:

| Parameter | Value | Description |
|-----------------------|-----------------|---|
| Source Trunk Group IP | <trunkID> | Trunk ID for selected FXS port |
| Destination IP Group | <BT IP Group> | IP Group for Business Talk aSBC |
| SIP Interface | <SIP Interface> | SIP Interface for Business Talk aSBC access |

Go to **SETUP > SIGNALING & MEDIA > GATEWAY > Routing > IP->Tel Routing**. Create new entry:

| Parameter | Value | Description |
|---------------------------|------------------|---|
| Source SIP Interface | <SIP Interface> | SIP Interface for Business Talk aSBC access |
| Destination Phone Pattern | <FAX DID> | Set FAX DID accessed by BT |
| Destination Type | Trunk Group | |
| Trunk Group ID | <Trunk Group IP> | Trunk ID for selected FXS port |
| Source IP Group | <BT IP Group> | IP Group for Business Talk aSBC |

Go to **SETUP > SIGNALING & MEDIA > SBC > Routing > IP-to-IP Routing**. Create new entry:

| Parameter | Value | Description |
|------------------------------|---------------|---------------------------------|
| Source IP Group | <BT IP Group> | IP Group for Business Talk aSBC |
| Destination Username Pattern | <FAX DID> | Set FAX DID accessed by BT |
| Destination Type | Gateway | |

When created please move new entry before default Business Talk route.



➤ V34-fax-transport-type

The next, V34FaxTransportType parameter can be set only using CLI/configuration file and is not visible in web application. To set this parameter go to dedicated configuration page: <https://<MediantIP>/AdminPage> (note: subpage address is case sensitive).

Go to “ini Parameters” subsite using left sided menu.

Parameter name: **V34FAXTRANSPORTTYPE**

Enter value: **1**

Click “Apply New Value”.

If parameter is set correctly you should see output:

Parameter Name: V34FAXTRANSPORTTYPE
Parameter New Value: 1
Parameter Description:Determines the V.34 fax transport method.

4.11.2 FXS fax on MediaPack cascaded behind Mediant

The fax integration on MediaPack with Business Talk through Mediant is based on assumption that fax calls are not sent to Mediation Server. In such scenario Mediant gateway only mediates in communication.

➤ MediaPack configuration

The MediaPack gateway must be first integrated directly with Mediant. The MediaPack endpoints are registered to Mediant using SIP REGISTER

Telephony Profile

The telephony profile assigned to FXS port must be updated to enable T.38 protocol. Go to **VoIP -> Coders and Profiles -> Tel Profile Settings**. Select appropriate profile (or create new one) and update **Fax Signaling Method** to **T.38 Relay**:

Note: Assigned Tel Profile can be checked under **VoIP -> GW and IP to IP -> Hunt Group -> Endpoint Phone Number**

Configure fax transmission parameters

Go to **VoIP -> Media -> Fax/Modem/CID Settings** and set following parameters:

| Parameter | Value | Description |
|----------------------------|-------------------|--------------------------------------|
| Fax Transport Mode | T.38 Relay | Enable T.38 |
| V.34 Modem Transport Type | Disable | Disable V.34 signals (block SG3 fax) |
| Fax Relay Redundancy Depth | 1 | Redundancy of transmitting pages |

| | | |
|-------------------------------------|---|---------------------------------|
| Fax Relay Enhanced Redundancy Depth | 4 | Redundancy of fax signalization |
|-------------------------------------|---|---------------------------------|

➤ Mediant configuration

Configuration starts from integration with MediaPack.

IP to IP Routing

Click **New** to create routing for outgoing fax calls from MediaPack to BT/BTIP

| Parameter | Value | Description |
|------------------------------------|--------------------|---|
| General > Name | MediaPack_AD_to_BT | |
| Match > Source IP Group | IPG_MediaPack_AD | |
| Match > Request Type | All | |
| Action > Destination Type | IP Group | |
| Action > Destination IP Group | <BT IP Group> | IP Group for Business Talk aSBC |
| Action > Destination SIP Interface | <SIP Interface> | SIP Interface for Business Talk aSBC access |

Click **New** to create routing for incoming fax calls from BT/BTIP to MediaPack

| Parameter | Value | Description |
|------------------------------|--------------------|-------------|
| General > Name | BT_to_MediaPack_AD | |
| Match > Source IP Group | <BT IP Group> | |
| Match > Request Type | All | |
| Match > Destination Username | <Fax phone number> | |
| Action > Destination Type | All Users | |

Note: place these rules before default entry forwarding calls to Microsoft

Also, calls must be routed directly:

From IP Group defined for calls from MediaPack towards Business Talk

From IP Group defined for calls from Business Talk towards "All Users" destination (if MediaPack is configured to register FXSW ports on Mediant)

5 Ribbon SBC Configuration Checklist for BTIP/BTalk/BToI/BTIPOI

The checklist below presents all steps of configuration required for VISIT SIP Teams offer deployment.

5.1 Flow matrix with BToI

| Source | Source IP | Source port | Destination | Destination IP | Destination port | Comment |
|-------------|-----------------|---|-------------|-----------------|---|-------------------|
| SBC | SBC public @IP | TLS 5061 | BToI | BToI public @IP | TLS 5061 | SIP Outgoing |
| BToI | BToI public @IP | Any | SBC | SBC public @IP | TLS 5061 | SIP Incoming |
| SBC | SBC public @IP | Range defined in Media System Configuration | BToI | BToI public @IP | UDP 6000-20000 | Media outgoing |
| BToI | BToI public @IP | UDP 6000-20000 | SBC | SBC public @IP | Range defined in Media System Configuration | Media incoming |

5.2 Flow matrix with BTIPOI

| Source | Source IP | Source port | Destination | Destination IP | Destination port | Comment |
|---------------|-------------------------|---|---------------|-------------------------|---|-------------------|
| SBC | SBC public @IP | TLS 5061 | BTIPOI | BTIPOI SIP public @IP | TLS 5061 | SIP Outgoing |
| BTIPOI | BTIPOI SIP public @IP | Any | SBC | SBC public @IP | TLS 5061 | SIP Incoming |
| SBC | SBC public @IP | Range defined in Media System Configuration | BTIPOI | BTIPOI media public @IP | UDP 6000-38000 | Media outgoing |
| BTIPOI | BTIPOI media public @IP | UDP 6000-38000 | SBC | SBC public @IP | Range defined in Media System Configuration | Media incoming |



5.3 Configuration checklist for Office365 Tenant

Parameters to configure the SIP Trunk between Tenant and SBC:

| | |
|-----------------------|---|
| Fqdn | <i><Customer SBC Public FQDN></i> |
| SipSignallingPort | <i>5062</i> |
| MaxConcurrentSessions | <i><Number of max sessions></i> |
| Enabled | <i>\$true</i> |
| ForwardPai | <i>\$true</i> |
| SendSipOptions | <i>\$true</i> |

Powershell cmdlet to add a new SBC :

```
#New-CsOnlinePSTNGateway -Fqdn sbc.contoso.com -SipSignallingPort 5062 -
MaxConcurrentSessions 50 -Enabled $true -ForwardPai $true -MediaBypass $false -
ForwardCallHistory $true -SendSipOptions $true
```

Powershell cmdlet to update an SBC on Tenant (without MBP)

```
Set-CsOnlinePSTNGateway -Identity sbc.contoso.com -SipSignallingPort 5062 -
MaxConcurrentSessions 50 -Enabled $true -ForwardPai $true -MediaBypass $false -
ForwardCallHistory $true -SendSipOptions $true
```

To activate media bypass:

```
#Set-CsOnlinePSTNGateway -Identity sbc.contoso.com -MediaBypass $true
```

To activate Local Media Optimization Europe:

```
Set-CsOnlinePSTNGateway -Identity sbc.contoso.com -GatewaySiteID $null -
MediaBypass $true -ProxySBC $null -BypassMode Always
```


5.4 Configuration checklist for QoS in Teams client

QoS management is done by configuring the Teams.exe at Windows level.

This configuration is done either locally or by GPO:

- Locally: Use policy-based Quality of Service (QoS) within Group Policy, and create a policy for Teams Audio with following parameters:

| Parameter | Value | Description |
|--------------------------|-------------|---|
| Policy Name | Teams Audio | |
| Application Name | Teams.exe | |
| Protocol | Both | TCP and UDP |
| Source Port Start | 50000 | Source ports used by Teams desktop clients are managed in the Teams Admin center. Microsoft recommends to keep this initial port range. |
| Source Port End | 50019 | |
| DSCP value | 46 | DSCP=46 Expedited Forwarding (EF) |

- By GPO: `#new-NetQosPolicy -Name "Teams Audio" -AppPathNameMatchCondition "Teams.exe" -IPProtocolMatchCondition Both -IPSrcPortStartMatchCondition 50000 -IPSrcPortEndMatchCondition 50019 -DSCPAction 46 -NetworkProfile All`

Configuration checklist regarding **Ribbon SBC – Standalone**

5.5 Step 1 – Teams Configuration

Certificate for the SBC Direct Routing Interface

On the SBC WebUi Interface:
Settings > Security > SBC Certificates

- Generate SBC Edge CSR
- Common name: **@hostname.domain.tld**
 - ISO Country code: **@Country**
 - Locality: **@Locality**
 - Organization: **@Organization**
 - Key Length: **2048**

TLS Profile Teams Side

On the SBC WebUi Interface:
Security > TLS Profiles

- Create TLS Profile Teams
- Description: **TLS Profile Teams**
 - TLS Protocol: **TLS 1.2 Only**
 - Mutual Authentication: **Disabled**
 - Verify Peer Server Certificate: **Disabled**

Node Level Settings

On the SBC WebUi Interface:
System > Node-Level Settings

- Configure DNS and NTP with your appropriate configuration
- Hostname: **@hostname**
 - Domain Name: **@domain.tld**

 - Time Zone: **GMT+1**
 - Use NTP: **Yes**
 - NTP Server: **@NTP_IPAddress**

 - Use Primary DNS: **Yes**
 - Primary Server IP: **@DNS_IPAddress**
 - Use Secondary DNS: **Yes**
 - Secondary Server IP: **@DNS_IPAddress**

Node Interface (1000/2000) / Networking Interfaces (SWe-Lite)

On the SBC WebUi Interface:

For SWe-Lite
Networking Interfaces > Logical Interfaces

For 1000/2000
Node Interfaces > Logical Interfaces

- Configure the parameters
- Ethernet 1 IP
 - Description: **Interface-Teams and BTol/BTIPol**
 - Primary Address: **@Private-IPAddress-BT(all) - Teams**
 - Primary Netmask: **@Netmask**
 - Media Next Hop IP: **@Gateway** (Swe-Lite only)

IF BT / BTIP :

This interface is dedicated to Teams Only
Node Interface Ethernet 2 IP is for BT/BTIP

IF BTol / BTIPol :

This interface is shared between Teams and BTol/BTIPol
Node Interface Ethernet 2 IP should be disabled

SIP Profile

On the SBC WebUi Interface:
SIP > SIP Profiles

- Create SIP Profile Teams
- Description : **SIP Profile Teams**
 - FQDN in From Header Location: **SBC Edge FQDN**
 - FQDN in Contact Header: **SBC FQDN**
 - Calling info source: **“From” Header Only**



| Media SDES-SRTP Profile | |
|---|---|
| On the SBC WebUi Interface: Media > SDES-SRTP Profiles | Create new Media SDES-SRTP Profile <ul style="list-style-type: none"> - Description: MCP Teams - Operation Option: Required |
| Media Profile | |
| On the SBC WebUi Interface: Media > Media Profiles | Create Media Profile > Voice Codec Profile <ul style="list-style-type: none"> - Description : Default G711A - Codec: G.711 A-law Create Media Profile > Voice Codec Profile <ul style="list-style-type: none"> - Description : Default G711u - Codec: G.711 u-law |
| Media List | |
| On the SBC WebUi Interface: Media >Media List | Create new Media List: <ul style="list-style-type: none"> - Description : ML Teams - Media Profil List: Default G711A Default G711u - SDES-SRTP Profile : MCP Teams - Media DSCP: 46 |
| SIP Server Table | |
| On the SBC WebUi Interface: SIP > SIP Server Tables | Create new SIP Server Table: <ul style="list-style-type: none"> - Description: SIP Server Teams Create SIP Server IP/FQDN Server Host <ul style="list-style-type: none"> - HOST FQDN/IP: sip.pstnhub.microsoft.com - Port: 5061 - Protocol: TLS - TLS Profile: TLS Profile Teams Transport <ul style="list-style-type: none"> - Monitor: SIP Options - Keep Alive Frequency: 180s Create SIP Server IP/FQDN Server Host <ul style="list-style-type: none"> - HOST FQDN/IP: sip2.pstnhub.microsoft.com - Port: 5061 - Protocol: TLS - TLS Profile: TLS Profile Teams Transport <ul style="list-style-type: none"> - Monitor: SIP Options - Keep Alive Frequency: 180s Create SIP Server IP/FQDN Server Host <ul style="list-style-type: none"> - HOST FQDN/IP: sip3.pstnhub.microsoft.com - Port: 5061 - Protocol: TLS - TLS Profile: TLS Profile Teams Transport <ul style="list-style-type: none"> - Monitor: SIP Options - Keep Alive Frequency: 180s |

| Voice Routing | |
|---|---|
| <p>On the SBC WebUi Interface: SIP > Transformation Tables</p> <p>Call Routing > Transformation</p> <p>Call Routing > Call Routing Table</p> | <p>Create new Transformation Table:</p> <ul style="list-style-type: none"> - Description: Teams to BTIP (or BTOI/BTIPol) <p>Create new Call Routing Table:</p> <ul style="list-style-type: none"> - Description: CR Teams to BTIP (or BTOI/BTIPol) <p>Update CR Teams to BTIP (or BTOI/BTIPol)</p> <ul style="list-style-type: none"> - Description: Teams to BTIP (or BTOI/BTIPol) - Number/Name transformation Table: Teams to BTIP (or BTOI/BTIPol) - Destination Signaling Group: SG to BT / BTIP (or BTOI/BTIPol) * refer to BTIP Configuration |
| Message Manipulation | |
| <p>SIP > Message Manipulation > Condition Rules Table</p> <p>SIP > Message Manipulation > Message Rules Table</p> | <p>Create new Condition Rule:</p> <ul style="list-style-type: none"> - Description: Anonymous Call Checking - Match Type: from - Operation: Regex - Match Regex: ^(?!anonymous).*\$ <p>Create new Condition Rule:</p> <ul style="list-style-type: none"> - Description: URL User - Match Type: SG User Value 1 - Operation: Regex - Match Regex: (< sip:)(.*)@(.*) <p>Create new SIP Message Rule Table:</p> <ul style="list-style-type: none"> - Description: Privacy-Removal - Application message: Selected Message - Message Selection: INVITE <p>Create new Header Rule:</p> <ul style="list-style-type: none"> - Description: Privacy Removal - Condition expression: Match All Condition > Anonymous Call Checking - Header Action: Remove - Header Name: Privacy <p>Create new SIP Message Rule Table:</p> <ul style="list-style-type: none"> - Description: Privacy-Modification - Application message: Selected Message - Message Selection: INVITE <p>Create new Header Rule:</p> <ul style="list-style-type: none"> - Description: Privacy Modification - Header Action: Modify - Header Name: Privacy - Header Value – Modify: id (Literal) <p><u>For BTIPol only:</u> Create new SIP Message Rule Table:</p> <ul style="list-style-type: none"> - Description: Diversion to Hsitory-Info twds Teams |

| | |
|---|--|
| <p>SIP > Message Manipulation > Message Rules Table</p> | <ul style="list-style-type: none"> - (Note that Hsitory is miswritten intentionally) - Application message: All Messages - Message Selection: INVITE <p>Create new Header Rule:</p> <ul style="list-style-type: none"> - Description: Collect Diversion - Header Action: Modify - Header Name: Diversion - Header value: SG User Value 1 - <p>Create new Header Rule:</p> <ul style="list-style-type: none"> - Description: Send Diversion to History-Info - Condition Expression: URL User - Header Action: Add - Header Name: History-Info - Header value: SG User Value 1 <p>Create new Header Rule:</p> <ul style="list-style-type: none"> - Description: Remove Diversion - Header Action: Remove - Header Name: Diversion <p>Create new Condition Rule:</p> <ul style="list-style-type: none"> - Description: Check Reason cause=34 - Match Type: SG User Value 2 - Operation: Regex - Match Regex: ^.*cause=34.*\$ <p>Create new SIP Message Rule Table:</p> <ul style="list-style-type: none"> - Description: Remove Reason header if cause=34 - Application message: Selected Message - Message Selection: 486 Busy Here <p>Create new Header Rule:</p> <ul style="list-style-type: none"> - Description: Detect Reason header cause=34 - Header Action: Modify - Header Name: Reason - Header Ordinal Number: All - Header Value: Copy Value to SG UserValue 2 <p>Create new Header Rule:</p> <ul style="list-style-type: none"> - Description: Remove Reason header if cause=34 - Condition Expression: Check Reason cause=34 - Header Action: Remove - Header Name: Reason - Header Ordinal Number: All - Header Value: Copy Value to SG UserValue 2 |
| <p>Signaling Group</p> | |
| <p>On the SBC WebUi Interface: Signaling Groups</p> | <p>Create new Signaling Group:</p> <ul style="list-style-type: none"> - Description: SG to Teams SIP Channels and Routing |

- Call Routing Table: **CR Teams to BTIP (or BTOI/BTIPoI)**
- SIP Profile: **SIP Profile Teams**
- SIP Server Table: **SIP Server Teams**
- Loadbalancing: **Priority: Register All**

Media Information

- Media List ID: **ML Teams**

*** if Media Bypass or Local Media Optimisation**

- RTCP Multiplexing: **Enable**

SIP IP Details

- Signaling/Media Private IP: **@Private-IPAddress-TeamsInterface**
- Signaling DSCP: **40**

*** if Local Media Optimization**

- Teams Local Media Optimization = Enable
- Private Media Source IP: **@Private-IPAddress-Local Interface.**

*** if Media Bypass or Local Media Optimization**

- ICE Support: Enable
- ICE Mode = Lite

*** if use NAT Outbound**

- Outbound NAT Traversal: **Static NAT**
- NAT Public IP: **@Public-IPAddress - TeamsInterface**

Listen Port

- Port: **5062**
- Protocol: **TLS**
- TLS Profile ID: **TLS Profile Teams**

Federated IP/FQDN

- IP/FQDN: **52.112.0.0**
- Netmask/prefix: **255.252.0.0**

- IP/FQDN: **52.120.0.0**
- Netmask/prefix: **255.252.0.0**

Message Manipulation: **Enabled**

Inbound Message Manipulation

Message Table List: **Privacy-Removal**
Remove Reason header if cause=34

Outbound Message Manipulation

Message Table List: **Privacy-Modification**
Diversion to Hsitory-Info twds Teams

5.6 Step 2 – BT / BTIP / BTOI / BTIPOI Configuration

Node Interface BT/BTIP only



| | |
|---|--|
| On the SBC WebUi Interface: Networking Interfaces > Logical Interfaces | Configure the parameters <ul style="list-style-type: none"> - Ethernet 2 IP - Description: Interface-BT/BTIP - Primary Address: @Private-IPAddress-BT/BTIP - Primary Netmask: @Netmask - Media Next Hop IP: @Gateway (Swe-Lite only) |
| SIP Profile | |
| On the SBC WebUi Interface: SIP > SIP Profiles | Create SIP Profile Teams <ul style="list-style-type: none"> - Description : SIP Profile BT/BTIP/BTOI - SBC Edge Diagnostic Header: Disable - UA Header : empty |
| TLS Profile BTol/BTIPol only | |
| On the SBC WebUi Interface: Security > TLS Profiles | Create TLS Profile for BTol/BTIPol <ul style="list-style-type: none"> - Description: BTol/BTIPol TLS Profile - TLS Protocol : TLS 1.2 Only - Mutual Authentication : Enabled - Client Cipher List : <ul style="list-style-type: none"> ▪ TLS_ECDHE_RSA_WITH_AES_256_CBC_SHA384 ▪ TLS_ECDHE_RSA_WITH_AES_128_CBC_SHA256 - Validation Client FQDN : Disabled |
| Media Profile | |
| On the SBC WebUi Interface: Media > Media Profiles | Create Media Profile > Voice Codec Profile <ul style="list-style-type: none"> - Description : Default G711A - Codec: G.711 A-law Create Media Profile > Voice Codec Profile <ul style="list-style-type: none"> - Description : Default G711μ - Codec: G.711 μ-law |
| SDES-SRTP Profiles BTol/BTIPol only | |
| On the SBC WebUi Interface: Media > SDES-SRTP Profiles | Edit the Teams SDES-SRTP profile: <ul style="list-style-type: none"> - Description : MCP Teams & BTOI |
| Media List for BTIP/BTIPOI | |
| On the SBC WebUi Interface: Media >Media List | Create new Media List: <ul style="list-style-type: none"> - Description : ML BT/BTIP, or ML BTOI/BTIPOI - Media Profil List: Default G711A - SDES-SRTP Profile : None (BTIP), or MCP Teams & BTOI (BTIPOI) - Media DSCP: 46 |
| Media List for BT/BTOI | |
| On the SBC WebUi Interface: Media >Media List | Create new Media List: <ul style="list-style-type: none"> - Description : ML BT/BTIP, or ML BTOI/BTIPOI - Media Profil List: Default G711A Default G711μ - SDES-SRTP Profile : None (BT), or MCP Teams & BTOI (BTOI) - Media DSCP: 46 |
| SIP Server Table BTol | |
| On the SBC WebUi Interface: SIP > SIP Server Tables | Create new SIP Server Table: <ul style="list-style-type: none"> - Description: SIP Server BTol Create SIP Server IP/FQDN Server Host |

| | |
|---|--|
| | <ul style="list-style-type: none"> - Priority: 1 - HOST FQDN/IP: Nominal-SBC-BToI-FQDN - Port: 5061 - Protocol: TLS - TLS Profile: TLS Profile Teams & BToI / BTIPoI - Monitor: SIP Options - Keep Alive Frequency: 300s <p>Create SIP Server IP/FQDN</p> <p>Server Host</p> <ul style="list-style-type: none"> - Priority: 2 - HOST FQDN/IP: Backup-SBC-BToI-FQDN - Port: 5061 - Protocol: TLS - TLS Profile: TLS Profile Teams & BToI / BTIPoI - Monitor: SIP Options - Keep Alive Frequency: 300s |
| SIP Server Table BTIPoI | |
| <p>On the SBC WebUi Interface: SIP > SIP Server Tables</p> | <p>Create new SIP Server Table:</p> <ul style="list-style-type: none"> - Description: SIP Server BTIPoI <p>Create SIP Server SRV</p> <p>Server Host</p> <ul style="list-style-type: none"> - HOST SRV : SRV SBC-BTIPoI - Port: 5061 - Protocol: TLS - TLS Profile: TLS Profile Teams & BToI / BTIPoI - Monitor: SIP Options - Keep Alive Frequency: 300s |
| SIP Server Table BT/BTIP | |
| <p>On the SBC WebUi Interface: SIP > SIP Server Tables</p> | <p>Create new SIP Server Table:</p> <ul style="list-style-type: none"> - Description: SIP Server BT/BTIP <p>Create SIP Server IP/FQDN</p> <p>Server Host</p> <ul style="list-style-type: none"> - Priority: 1 - HOST FQDN/IP: @SBC-BT/BTIP-IPAddress - Port: 5060 - Protocol: TCP <p>Transport</p> <ul style="list-style-type: none"> - Monitor: SIP Options - Keep Alive Frequency: 300s <p>Create SIP Server IP/FQDN</p> <p>Server Host</p> <ul style="list-style-type: none"> - Priority: 2 - HOST FQDN/IP: @SBC-BT/BTIP-IPAddress - Port: 5060 - Protocol: TCP <p>Transport</p> <ul style="list-style-type: none"> - Monitor: SIP Options - Keep Alive Frequency: 300s |



| Voice Routing | |
|---|---|
| <p>On the SBC WebUi Interface: SIP > Transformation Tables</p> <p>Call Routing > Call Routing Table</p> | <p>Create new Transformation Table:</p> <ul style="list-style-type: none"> - Description: BT/BTIP (or BTOI/BTIPol) to Teams <p>Create new Call Routing Table:</p> <ul style="list-style-type: none"> - Description: CR BT/BTIP (or BTOI/BTIPol) to Teams <p>Update CR BT/BTIP (or BTOI/BTIPol) to Teams</p> <ul style="list-style-type: none"> - Description: BT/BTIP (or BTOI/BTIPol) to Teams - Number/Name transformation Table: BT/BTIP (or BTOI/BTIPol) to Teams - Destination Signaling Group: SG to Teams* <p><i>* refer to Teams Configuration</i></p> |
| Message Manipulation | |
| <p>On the SBC WebUi Interface: SIP > Message Manipulation > Message Rules Table</p> | <p>Create new SIP Message Rule Table:</p> <ul style="list-style-type: none"> - Description: User-Agent <p>Create new Header Rule:</p> <ul style="list-style-type: none"> - Description: User-Agent - Header Action: Modify - Header Name: User-Agent - Header Value: Modify - Add/Edit: <ul style="list-style-type: none"> o Type of value: Token o Value: user-agent o Suffix: \ Teams |
| Signaling Group BT/BTIP | |
| <p>On the SBC WebUi Interface: Signaling Groups</p> | <p>Create new Signaling Group:</p> <ul style="list-style-type: none"> - Description: SG to BT / BTIP <p>SIP Channels and Routing</p> <ul style="list-style-type: none"> - Call Routing Table: CR BT - All to Teams - SIP Profile: SIP Profile BT - SIP Server Table: SIP Server BT – All - Load balancing: Priority: Register All <p>Media Information</p> <ul style="list-style-type: none"> - Media List ID: ML BT/BTIP <p>SIP IP Details</p> <ul style="list-style-type: none"> - Signaling/Media Private IP: Ethernet 2 IP - Signaling DSCP: 40 <p>Listen Port</p> <ul style="list-style-type: none"> - Port: 5060 - Protocol: TCP <p>Federated IP/FQDN</p> <ul style="list-style-type: none"> - IP/FQDN: @SBC-BT/BTIP-IPAddress - Netmask/prefix: 255.255.255.255 - IP/FQDN: @SBC-BT/BTIPI-IPAddress |

| | |
|---|---|
| | <ul style="list-style-type: none"> - Netmask/prefix: 255.255.255.255 <p>Message Manipulation: Enabled</p> <p>Outbound Message Manipulation</p> <ul style="list-style-type: none"> - Message Table List: User-Agent |
| Signaling Group BTol / BTIPol | |
| <p>On the SBC WebUi Interface: Signaling Groups</p> | <p>Create new Signaling Group:</p> <ul style="list-style-type: none"> - Description: SG to BTol / BTIPol <p>SIP Channels and Routing</p> <ul style="list-style-type: none"> - Call Routing Table: CR BTol / BTIPol to Teams - SIP Profile: SIP Profile BTol / BTIPol - SIP Server Table: SIP Server BTol / BTIPol - All - For BTol, Load balancing: Priority: Register All <p>Media Information</p> <ul style="list-style-type: none"> - Media List ID: ML BTol/BTIPol <p>SIP IP Details</p> <ul style="list-style-type: none"> - Signaling/Media Private IP: Ethernet 1 IP - Signaling DSCP: 40 - Outbound NAT Traversal : None (General recommendation is not to use NAT translation for BTol/BTIPol. The SBC IP interface dedicated to Teams and BTol/BTIPol will be assigned the SBC Public IP address). <p>Listen Port</p> <ul style="list-style-type: none"> - Port: 5061 - Protocol: TLS - TLS Profile: BTol/BTIPol TLS profile <p>Federated IP/FQDN</p> <ul style="list-style-type: none"> - IP/FQDN: FQDN SBC-BTOI / BTIPOI (IP acceptable for BTol) - Netmask/prefix: 255.255.255.255 - IP/FQDN: FQDN SBC-BTOI / BTIPOI (IP acceptable for BTol) - Netmask/prefix: 255.255.255.255 <p>Message Manipulation: Enabled</p> <p>Outbound Message Manipulation</p> <ul style="list-style-type: none"> - Message Table List: User-Agent |

5.7 Ribbon FAX configuration checklist

5.7.1 FXS fax with Ribbon configuration

The following guide describes steps which should be followed to enable the use of analogue fax devices on Ribbon Gateway. It is assumed that initial configuration of the Ribbon gateway is already done.

5.7.2 Media Profile

It is necessary to enable T.38 support by setting T.38 Fax as a codec in Media Profile tab. In order to do that go to **SETTINGS > MEDIA > MEDIA PROFILE**

Create a new profile by pressing **Create Media Profile** and then **Fax Codec Profile**

| Parameter | Value | Description |
|------------------------------|--------------------|--|
| Description | T38 Profile | Profile name |
| Codec | T.38 Fax | Select T.38 protocol for fax transmission |
| Signalling Packet Redundancy | 4 | Signalization redundancy |
| Payload Packet Redundancy | 1 | Page transmission redundancy |
| Fallback to Passthrough | Disabled | FAX transmission cannot fallback to G711 passthrough. BT does not support G711 passthrough mode |
| Super G3 to G3 Fallback | Enabled | Force SG3 Fax calls switch to G3 mode. Speed is reduced to 14400bps. ECM is not disabled administratively. |

5.7.3 Fax Media List

Go to **SETTINGS > MEDIA > MEDIA LIST** and press  to add a new Media List.

| Parameter | Value | Description |
|-------------------------|---------------------------------------|--|
| Description | FAX Media List | Media List name |
| Media Profiles List | Default G711A T.38 Profile | Add here the voice codec (here: G.711A) and the fax media codec (here: T.38 Profile) |
| Digit (DTMF) Relay Type | RFC 2833 | Specifies how DTMF digits are passed through data network. |
| Modem Passthrough | Disabled | Specifies whether modem passthrough is enabled when using the G.711 codec. |
| Fax Passthrough | Disabled | Specifies whether fax passthrough is enabled when using the G.711 codec. |
| CNG Tone Detection | Disabled | Specifies whether the SONUS-SBC system will detect Fax tones produced by the origination side fax machine. |

5.7.4 FXS port configuration

To configure an FXS port go to **SETTINGS > NODE INTERFACES** and select the port to which a Fax machine will be connected.

| Parameter | Value | Description |
|---------------------|------------------------|-------------------------------|
| Analog Line Profile | <Country> | A country dependent parameter |

5.7.5 CAS Signalling Profile

CAS Signalling Profiles control various aspects of loop start, DTMF, tone detection and other features associated with the variants of CAS calls. In order to create a CAS Signaling Profile go to **SETTINGS > CAS > CAS SIGNALING PROFILES**

Create a new profile by selecting **Create CAS Profile** and then **FXS Profile**.



| Parameter | Value | Description |
|-----------------|----------------|---------------------------------|
| Description | <Profile Name> | CAS Signalling Profile name |
| Loop Start Type | Basic | Specifies the Loop Start method |

5.7.6 Transformation Table

FXS FAX Towards BT

Outgoing FXS Fax makes use of the same Transformation Table as standard outgoing BT calls

BT Towards FXS FAX

Create a new Transformation Table for faxes incoming from BT. Go to **SETTINGS > CALL ROUTING > TRANSFORMATION** then press  and fill the description field to name the table. Select the newly created table and press  to add a new entry.

| Parameter | Value | Description |
|----------------------|-----------------------------|--|
| Match Type | Mandatory / Optional | This option states whether the number matching should be mandatory or optional |
| Input Field - Value | <FXS Fax number> | Number matching rule. The backslash is used to treat plus "+" as character and not regex special symbol. |
| Output Field - Value | < FXS Fax number> | Set the same number in transformation output. |

5.7.7 CAS Signalling Group

New CAS signalling group for fax devices must be created on Ribbon gateway. Calls from CAS dedicated for faxes will be routed differently so existing CAS for analogue phones cannot be used.

| Parameter | Value | Description |
|-----------------------|---------------------------------|---|
| Description | <CAS Signaling Group Name> | CAS Signalling Group Name |
| Channel Hunting | Own Number | Parameter must be set to Own Number to send incoming calls to a proper fax machine |
| Call Routing Table | <Call Routing Table Towards BT> | Select existing Call Routing Table Towards BT |
| CAS Signaling Profile | <CAS Signaling Profile> | Select existing CAS Signalling Profile |

In **Assigned Channels** table create a new entry with dedicated phone number for each fax port.




5.7.8 Call Routing Table

FXS FAX Towards BT

Outgoing FXS Fax makes use of the same Call Routing Table as standard outgoing BT calls

BT Towards FXS FAX

Go to **SETTINGS > CALL ROUTING > CALL ROUTING TABLE** and select a proper call routing table for outgoing calls towards BT. Afterwards press  to add an entry to the table.

| Parameter | Value | Description |
|-------------------------------------|--|---|
| Number/Name Transformation Table | <Transformation Table BT Towards FXS fax> | Select proper Transformation Table for incoming FXS fax |
| Destination Signaling Groups | <CAS Signaling Group> | Select existing CAS FXS Signalling Group |
| Media Mode | DSP | Enable Ribbon DSP resources for FAX transcoding purpose |
| Media List | FAX Media List | Select media list containing T.38 codec. |

5.7.9 Update Codecs

Please make sure that FAX Media List is configured on the following:

Call Routing Table entry from CAS (FXS FAX) to BT

Call Routing Table entry from BT to CAS (FXS FAX)

Business Talk SIP Signaling Group(s)



6 Oracle SBC Configuration Checklist for BTIP/BTalk

6.1 Warning

This configuration checklist is for S-Cz8.4.0 and upper releases. If checklist for S-Cz8.3.0 is required, please ask your sale contact.

6.2 Configuration Requirements

6.2.1 Tenant configuration

Pair the SBC to the tenant:

Command:

```

New-CsOnlinePSTNGateway
-Fqdn <SBC FQDN>
-SipSignallingPort 5061
-MaxConcurrentSessions <Number>
-Enabled $true
-ForwardPai $true
-ForwardCallHistory $true
-SendSipOptions $true
-MediaBypass $true
-ProxySBC $null
-BypassMode Always

```

6.2.2 Firewall and used ports

With media-bypass mode activated, media flow do not more go through Microsoft Media-Processor equipment.

| Source | IP | Destination | IP | Protocol | Port | Comment |
|-----------------|---------------|-----------------|---------------|----------|-------------|---------------|
| SBC | Public @IP | Any | Any | UDP | 53 | Public DNS |
| SBC | Public @IP | Teams SIP Proxy | 52.114.0.0/16 | TCP | 5061 | SIP/TLS |
| Teams SIP Proxy | 52.114.0.0/16 | SBC | Public @IP | TCP | 5061 | SIP/TLS |
| Internet | Any | SBC | Public @IP | UDP | 49152-65535 | Media traffic |
| SBC | Public @IP | Internet | Any | UDP | 49152-53247 | Media traffic |

6.2.3 SBC network configuration (wired or virtually wired)

Five interfaces will need IP addresses:

- For the Internet facing LAN: 1 private IP NATed to the public IP (if NAT)
- Customer trunk LAN: 1 IP address
- [Optional] HA LAN: 1 or 2 IP address depending if HA is redounded
- Admin configuration: 1 IP address for CLI configuration

6.2.4 Certificates

Following requirements regarding Certificate configuration:

- Certificate of the certification authority, signing the Microsoft Phone System Direct Routing service (DigiCert Global Root G2, format X.509 Base64)
- 1 cyphered file containing both the private key and the public certificate per domain used on the SBC, signed by a trusted Certificate Authority to be known by Microsoft Phone System Direct Routing service, aka such as DigiCert CA which Orange has chosen as CA provider
Note: for the multi-tenant SBC solution, this should be a wildcard certificate, aka *.teams.orange.com
- Certificate of the trusted certificate authority, and of each sub-authority having signed the above certificate (format X.509 Base64)

6.2.5 License

- No license is required

6.2.6 User Agent

Within VISIT SIP Teams context, User agent header must have following format:

User-Agent: ORACLE <SBC Model>/v.8.3.0\ Teams

6.2.7 QoS in Teams Client

QoS management is done by configuring the Teams.exe at Windows level.

This configuration is done either locally or by GPO:

- Locally: Use policy-based Quality of Service (QoS) within Group Policy, and create a policy for Teams Audio with following parameters:



| Parameter | Value | Description |
|-------------------|-------------|---|
| Policy Name | Teams Audio | |
| Application Name | Teams.exe | |
| Protocol | Both | TCP and UDP |
| Source Port Start | 50000 | Source ports used by Teams desktop clients are managed in the Teams Admin center. Microsoft recommends to keep this initial port range. |
| Source Port End | 50019 | |
| DSCP value | 46 | DSCP=46 Expedited Forwarding (EF) |

- By GPO: `#new-NetQosPolicy -Name "Teams Audio" -AppPathNameMatchCondition "Teams.exe" -IPProtocolMatchCondition Both -IPSrcPortStartMatchCondition 50000 -IPSrcPortEndMatchCondition 50019 -DSCPAction 46 -NetworkProfile All`

6.3 ORACLE SBC - Standalone

6.3.1 First step

➤ Setup Entitlements

Configure SBC features:

| Element | Configuration |
|--------------------|---------------------------------------|
| Setup entitlements | OracleSBC1# setup entitlements |

Example:

```
OracleSBC1# setup entitlements

-----
Entitlements for Session Border Controller
Last Modified: 2019-04-23 09:58:38
-----
 1 : Session Capacity           : 12000
 2 : Accounting                 : enabled
 3 : IPv4 - IPv6 Interworking   :
 4 : IWF (SIP-H323)            : enabled
 5 : Load Balancing            : enabled
 6 : Policy Server              : enabled
 7 : Quality of Service        : enabled
 8 : Routing                   : enabled
 9 : SIPREC Session Recording  : enabled
10: Admin Security              :
11: ANSSI R226 Compliance      :
12: IMS-AKA Endpoints          : 0
13: IPSec Trunking Sessions    : 0
14: MSRP B2BUA Sessions       : 0
15: SRTP Sessions              : 2000
16: Transcode Codec AMR Capacity : 0
17: Transcode Codec AMRWB Capacity : 0
```

To “unlock” SRTP feature like “media-sec-policy” you must give a number for “SRTP Sessions”.

6.3.2 IP Network configuration

➤ Physical interface configuration

Three network interfaces:

- wancom0: admin interface (CLI)
- s0p0: media interface to Teams (media)
- s0p1: media interface to BTIP (media)

| Interface | Configuration |
|----------------|---|
| Wancom0 | Configured at first initial boot |
| S0p0 | <pre>OracleSBC1# configure terminal OracleSBC1 (configure)# system OracleSBC1 (system)# phy-interface OracleSBC1 (phy-interface)# name s0p0 OracleSBC1 (phy-interface)# operation-type Media OracleSBC1 (phy-interface)# port 0 OracleSBC1 (phy-interface)# slot 0 OracleSBC1 (phy-interface)# done</pre> |
| S0p1 | <pre>OracleSBC1# configure terminal OracleSBC1 (configure)# system OracleSBC1 (system)# phy-interface OracleSBC1 (phy-interface)# name s0p1 OracleSBC1 (phy-interface)# operation-type Media OracleSBC1 (phy-interface)# port 0 OracleSBC1 (phy-interface)# slot 1 OracleSBC1 (phy-interface)# done</pre> |

➤ Network interface configuration

| Interface | Configuration |
|--------------------------|--|
| Wancom0 | Configured at first initial boot |
| S0p0 To Teams | <pre>OracleSBC1# configure terminal OracleSBC1 (configure)# system OracleSBC1 (system)# network-interface OracleSBC1 (network-interface)# name s0p0 OracleSBC1 (network-interface)# hostname <SBC FQDN> OracleSBC1 (network-interface)# ip-address <SBC Public IP> OracleSBC1 (network-interface)# netmask <Netmask> OracleSBC1 (network-interface)# gateway <GW IP> OracleSBC1 (network-interface)# dns-ip-primary <Primary Public DNS> OracleSBC1 (network-interface)# dns-ip-backup1 <Backup Public DNS> OracleSBC1 (network-interface)# dns-domain <DNS Domain> OracleSBC1 (network-interface)# add-hip-ip <SBC Public IP> OracleSBC1 (network-interface)# add-icmp-ip <SBC Public IP></pre> |



| | |
|---------------------------------------|---|
| | <pre>OracleSBC1 (network-interface)# add-ssh-ip <SBC Public IP> OracleSBC1 (network-interface)# gw-heartbeat OracleSBC1 (gateway-heartbeat)# select OracleSBC1 (gateway-heartbeat)# state enabled OracleSBC1 (gateway-heartbeat)# heartbeat 1 OracleSBC1 (gateway-heartbeat)# retry-count 2 OracleSBC1 (gateway-heartbeat)# retry-timeout 1 OracleSBC1 (gateway-heartbeat)# health-score 30 OracleSBC1 (gateway-heartbeat)# done OracleSBC1 (gateway-heartbeat)# exit OracleSBC1 (network-interface)# done</pre> |
| <p>S0p1 To BTIP</p> | <pre>OracleSBC1# configure terminal OracleSBC1 (configure)# system OracleSBC1 (system)# network-interface OracleSBC1 (network-interface)# name s0p1 OracleSBC1 (network-interface)# ip-address <SBC Private IP> OracleSBC1 (network-interface)# netmask <Netmask> OracleSBC1 (network-interface)# gateway <GW IP> OracleSBC1 (network-interface)# add-hip-ip <SBC Private IP> OracleSBC1 (network-interface)# add-icmp-ip <SBC Private IP> OracleSBC1 (network-interface)# gw-heartbeat OracleSBC1 (gateway-heartbeat)# select OracleSBC1 (gateway-heartbeat)# state enabled OracleSBC1 (gateway-heartbeat)# heartbeat 1 OracleSBC1 (gateway-heartbeat)# retry-count 2 OracleSBC1 (gateway-heartbeat)# retry-timeout 1 OracleSBC1 (gateway-heartbeat)# health-score 31 OracleSBC1 (gateway-heartbeat)# done OracleSBC1 (gateway-heartbeat)# exit OracleSBC1 (network-interface)# done</pre> |



6.3.3 Teams configuration

➤ Certificate

Microsoft Teams Direct Routing Interface only allows TLS connections from SBCs for SIP traffic with a certificate signed by one of the trusted certification authorities.

The step below describes how to request a certificate for SBC External interface and configure it based on the example of Symantec | DigiCert.

The process includes the following steps:

1. Create a certificate-record “Certificate-record” is configuration element on Oracle SBC which captures information for a TLS certificate – such as common-name, key-size, key-usage etc. Following certificate-records are required on the Oracle ESBC in order for the SBC to connect with Microsoft Teams:

- SBC 1 certificate-record assigned to SBC
- IntermediateCA 1 certificate-record for intermediateCA
- Root 1 certificate-record for root cert (DigiCert Global Root G2 for Teams SIP proxy)

2. Generate a Certificate Signing Request (CSR) and obtain the certificate from a supported Certification Authority

3. Deploy the SBC and Root/Intermediary certificates on the SBC

Step 1 – Creating the certificate

Navigate to certificate-record config element under security and then configure a certificate record for SBC as shown below

| Element | Configuration |
|--------------------|---|
| Certificate-record | OracleSBC1# configure terminal |
| | OracleSBC1 (configure)# security |
| SBCCertificate | OracleSBC1 (security)# certificate-record |
| | OracleSBC1 (certificate-record)# name SBCCertificate |
| | OracleSBC1 (certificate-record)# country <Country> e.g. FR |
| | OracleSBC1 (certificate-record)# state <State> e.g. Paris |
| | OracleSBC1 (certificate-record)# locality <Locality> e.g. Paris |
| | OracleSBC1 (certificate-record)# organization <Organization> e.g. Orange |
| | OracleSBC1 (certificate-record)# unit <unit> e.g. “Orange Business Services” |
| | OracleSBC1 (certificate-record)# common-name <SBC FQDN> |
| | OracleSBC1 (certificate-record)# key-size 2048 |
| | OracleSBC1 (certificate-record)# trusted enabled |
| | OracleSBC1 (certificate-record)# key-usage-list (digitalSignature keyEncipherment) |
| | OracleSBC1 (certificate-record)# extended-key-usage-list (serverAuth ClientAuth) |
| | OracleSBC1 (certificate-record)# key-algor rsa |
| | OracleSBC1 (certificate-record)# digest-algor sha256 |
| | OracleSBC1 (certificate-record)# ecdsa-key-size p256 |
| | OracleSBC1 (certificate-record)# done |



| | |
|--|--|
| <p>Certificate-record</p> <p>IntermediateCA</p> | <pre> OracleSBC1# configure terminal OracleSBC1 (configure)# security OracleSBC1 (security)# certification-record OracleSBC1 (certificate-record)# name SBCInter OracleSBC1 (certificate-record)# country <Country> e.g. FR OracleSBC1 (certificate-record)# state <State> e.g. Paris OracleSBC1 (certificate-record)# locality <Locality> e.g. Paris OracleSBC1 (certificate-record)# organization <Organization> e.g. Orange OracleSBC1 (certificate-record)# unit <unit> e.g. "Orange Business Services" OracleSBC1 (certificate-record)# common-name SBCInter OracleSBC1 (certificate-record)# key-size 2048 OracleSBC1 (certificate-record)# trusted enabled OracleSBC1 (certificate-record)# key-usage-list (digitalSignature keyEncipherment) OracleSBC1 (certificate-record)# extended-key-usage-list (serverAuth ClientAuth) OracleSBC1 (certificate-record)# key-algor rsa OracleSBC1 (certificate-record)# digest-algor sha256 OracleSBC1 (certificate-record)# ecdsa-key-size p256 OracleSBC1 (certificate-record)# done </pre> |
| <p>Certificate-record</p> <p>DigiCert Global Root G2</p> | <pre> OracleSBC1# configure terminal OracleSBC1 (configure)# security OracleSBC1 (security)# certification-record OracleSBC1 (certificate-record)# name GlobalRootG2 OracleSBC1 (certificate-record)# country <Country> e.g. FR OracleSBC1 (certificate-record)# state <State> e.g. Paris OracleSBC1 (certificate-record)# locality <Locality> e.g. Paris OracleSBC1 (certificate-record)# organization <Organization> e.g. Orange OracleSBC1 (certificate-record)# unit <unit> e.g. "Orange Business Services" OracleSBC1 (certificate-record)# common-name "DigiCert Global Root G2" OracleSBC1 (certificate-record)# key-size 2048 OracleSBC1 (certificate-record)# trusted enabled OracleSBC1 (certificate-record)# key-usage-list (digitalSignature keyEncipherment) OracleSBC1 (certificate-record)# extended-key-usage-list serverAuth OracleSBC1 (certificate-record)# key-algor rsa OracleSBC1 (certificate-record)# digest-algor sha256 OracleSBC1 (certificate-record)# ecdsa-key-size p256 OracleSBC1 (certificate-record)# done </pre> |

Step 2 – Generating a certificate signing request

Generate a certificate signing request only for SBCCertificate to create a certificate request and upload it to DigiCert for signage:

| Element | Configuration |
|---|--|
| <p>Certificate signing request</p> | <pre> OracleSBC1# generate-certification-request SBCCertificate </pre> <p>Should return something like this :</p> |

```

|-----BEGIN CERTIFICATE REQUEST-----
MIIC7zCCAdcCAQAwYsxCzAJBgNVBAYTAkZSMQ4wDAYDVQQIEwVQYXJpczEOMAwG
A1UEBxMFUGFyaXMxDzANBgNVBAoTBk9yYW5nZTEhMB8GA1UECxMYT3Jhbmd1IEJ1
c2l1ZXNzIFN1cnZpY2VzMSgwJgYDVQDEEx9ocTYwNmdd3Lm1zdm1zaXQ2Lm9uZS51
cXVhbnQubmV0MIIBIjANBgkqhkiG9w0BAQEFAAOCAQ8AMIIBCgKCAQEAv/NxxJUw
2JeNE3NBvYfKYC0kniJkaB3r4Y+QCPo1gkL5qK9bsVTwop7mHmpon4N1gZQwfkqA
UigABDIWwiS0avplpo98FM5fCFg0J1GZXe/E9UReK9g4NV8D1yyDgZoeDWBtuq6h
S/n1ob96NjuH0+wSaxWvi0eQk0icB+7+o9YZ1tCS|           3gym7W4y7UKyJ
7b6FQ7DYJifsd6hAfASx9xfnDRW9RaWr1Qn1kjXeFXwJRSLLALaWn6vaFznAkXGL
5thaEi3MirbinRJQXdtmBqd5hF4wZxIQpZS4qVtVS00+EjQ7+mxvXyBQo08mzmSy
vSmWu3Yw3jZREwIDAQABoB4wHAYJKoZIhvcNAQkOMQ8wDTALBgNVHQ8EBAMCBaAw
DQYJKoZIhvcNAQELBQADggEBALZ1iCeQjm0ovxprVjF2NPnVEbYUcs2t8vJnCZr
1a6NxdBT5sF/100Fss67X/X8TAIg0Wx1Ii81xC7ydoItohpUtiII4R1zL6nJC5oP
brCHnqFRJbQxdnCFpWYDV3Rff8HsmicizHNv3cYbGyTWwbySuOpiA+RCTPST1Rg
vr1hdSVuCrzRCrt51nEE5X+Vmb0RK2nJ+4CGgNGy6MLyRQ0aIFxnRF/wCWdr+zjQ
rDjrRknTc0tB/QaQk1VgpcvZG3XJ90Q3toqXpo6F2Fq8q99/75aUg1680G8J3CMu
mZ5K+y1dScQuS6Dq1EJt0RZ8IUrNYcn8sB/uhGwXm4Qy79c=
-----END CERTIFICATE REQUEST-----

```

Step 3 – Deploy SBC & root/intermediate certificates

| | | | | | |
|----------------|---|-----------------|------------------|------------------------|------|
| SBCInter | → | IntermediateCA | 03/05/2019 12:06 | Certificat de sécur... | 2 Ko |
| SBCCertificate | → | ssl_certificate | 03/05/2019 12:06 | Certificat de sécur... | 3 Ko |

| Element | Configuration |
|--------------------|---|
| Import-certificate | OracleSBC1# import-certificate try-all SBCCertificate At this point – paste the signed SBC certificate and then issue command “;” |

➤ TLS Profile

Create a TLS context for Teams with following parameters:

TLS configuration needs as parameters the name of certificate records.

| Element | Configuration |
|---------|--|
| TLSv1.2 | OracleSBC1# configure terminal OracleSBC1 (configure)# security OracleSBC1 (security)# tls-profile OracleSBC1 (tls-profile)# name TeamsTLS OracleSBC1 (tls-profile)# end-entity-certificate <SBCCertificate> OracleSBC1 (tls-profile)# trusted-ca-certificates (<GlobalRootG2> <SBCInter>) OracleSBC1 (tls-profile)# cipher-list ALL OracleSBC1 (tls-profile)# verify-depth 10 OracleSBC1 (tls-profile)# mutual-authenticate enabled OracleSBC1 (tls-profile)# tls-version tlsv12 OracleSBC1 (tls-profile)# cert-status-check disabled OracleSBC1 (tls-profile)# ignore-dead-responder disabled |

| | |
|--|---|
| | <pre>OracleSBC1 (tls-profile)# allow-self-signed-cert disabled OracleSBC1 (tls-profile)# done</pre> |
|--|---|

➤ Enable media manager

| Element | Configuration |
|----------------------|--|
| Media-manager | <pre>OracleSBC1# configure terminal OracleSBC1 (configure)# media-manager OracleSBC1 (media-manager)# media-manager OracleSBC1 (media-manager-config)# state enabled OracleSBC1 (media-manager-config)# options +audio-allow-asymmetric-pt OracleSBC1 (media-manager-config)# options +xcode-gratuitous-rtcp-report-generation OracleSBC1 (media-manager-config)# options +dont-terminate-assoc-legs OracleSBC1 (media-manager-config)# done</pre> |

➤ Steering pool

| Element | Configuration |
|--------------------------------|--|
| Steering pool Teams | <pre>OracleSBC1# configure terminal OracleSBC1 (configure)# media-manager OracleSBC1 (media-manager)# steering pool OracleSBC1 (steering-pool)# ip-address <SBC Public IP> OracleSBC1 (steering-pool)# start-port 49152 OracleSBC1 (steering-pool)# end-port 65535 OracleSBC1 (steering-pool)# realm-id ToTeams OracleSBC1 (steering-pool)# done</pre> |
| Steering pool BTIP | <pre>OracleSBC1# configure terminal OracleSBC1 (configure)# media-manager OracleSBC1 (media-manager)# steering pool OracleSBC1 (steering-pool)# ip-address <SBC Private IP> OracleSBC1 (steering-pool)# start-port 6000 OracleSBC1 (steering-pool)# end-port 65535 OracleSBC1 (steering-pool)# realm-id ToBTIP OracleSBC1 (steering-pool)# done</pre> |

➤ Session agent

Microsoft Teams use different SIP Proxy who can send/receive traffic from the SBC.

| Element | Configuration |
|---|---|
| Session agent sip.pstnhub.microsoft.com | <pre>OracleSBC1# configure terminal OracleSBC1 (configure)# session-router OracleSBC1 (session-router)# session-agent OracleSBC1 (session-agent)# hostname sip.pstnhub.microsoft.com OracleSBC1 (session-agent)# port 5061</pre> |

| | |
|---|--|
| | <pre>OracleSBC1 (session-agent)# transport-method StaticTLS OracleSBC1 (session-agent)# realm-id ToTeams OracleSBC1 (session-agent)# ping-method OPTIONS OracleSBC1 (session-agent)# ping-interval 180 OracleSBC1 (session-agent)# refer-call-transfer enabled OracleSBC1 (session-agent)# done</pre> |
| <p>Session agent</p> <p>sip2.pstnhub.microsoft.com</p> | <pre>OracleSBC1# configure terminal OracleSBC1 (configure)# session-router OracleSBC1 (session-router)# session-agent OracleSBC1 (session-agent)# hostname sip2.pstnhub.microsoft.com OracleSBC1 (session-agent)# port 5061 OracleSBC1 (session-agent)# transport-method StaticTLS OracleSBC1 (session-agent)# realm-id ToTeams OracleSBC1 (session-agent)# ping-method OPTIONS OracleSBC1 (session-agent)# ping-interval 180 OracleSBC1 (session-agent)# refer-call-transfer enabled OracleSBC1 (session-agent)# done</pre> |
| <p>Session agent</p> <p>sip3.pstnhub.microsoft.com</p> | <pre>OracleSBC1# configure terminal OracleSBC1 (configure)# session-router OracleSBC1 (session-router)# session-agent OracleSBC1 (session-agent)# hostname sip3.pstnhub.microsoft.com OracleSBC1 (session-agent)# port 5061 OracleSBC1 (session-agent)# transport-method StaticTLS OracleSBC1 (session-agent)# realm-id ToTeams OracleSBC1 (session-agent)# ping-method OPTIONS OracleSBC1 (session-agent)# ping-interval 180 OracleSBC1 (session-agent)# refer-call-transfer enabled OracleSBC1 (session-agent)# done</pre> |

➤ Session group

Defined session group with all session agents configured earlier to prevent connectivity issue when the active session agent became unreachable:

| Element | Configuration |
|-----------------------------|--|
| <p>Session group</p> | <pre>OracleSBC1# configure terminal OracleSBC1 (configure)# session-router OracleSBC1 (session-router)# session-group OracleSBC1 (session-group)# group-name TeamsGrp OracleSBC1 (session-group)# dest (sip.pstnhub.microsoft.com sip2.pstnhub.microsoft.com sip3.pstnhub.microsoft.com) OracleSBC1 (session-group)# sag-recursion enabled OracleSBC1 (session-group)# stop-sag-recurse 300-407,409-599 OracleSBC1 (session-group)# done</pre> |

➤ Local policy

Create local policy to manage the traffic between different realm (BTIP to Teams / Teams to BTIP):

| Element | Configuration |
|--|---|
| Local policy From Teams to BTIP | OracleSBC1# configure terminal OracleSBC1 (configure)# session-router OracleSBC1 (session-router)# local-policy OracleSBC1 (local-policy)# from-address * OracleSBC1 (local-policy)# to-address * OracleSBC1 (local-policy)# source-realm ToTeams OracleSBC1 (local-policy)# policy-attribute OracleSBC1 (local-policy-attribute)# next-hop sag:BTIPGrp OracleSBC1 (local-policy-attribute)# realm ToBTIP OracleSBC1 (local-policy-attribute)# done OracleSBC1 (local-policy)# done |
| Local policy From BTIP to Teams | OracleSBC1# configure terminal OracleSBC1 (configure)# session-router OracleSBC1 (session-router)# local-policy OracleSBC1 (local-policy)# from-address * OracleSBC1 (local-policy)# to-address * OracleSBC1 (local-policy)# source-realm ToBTIP OracleSBC1 (local-policy)# policy-attribute OracleSBC1 (policy-attribute)# next-hop sag:TeamsGrp OracleSBC1 (policy-attribute)# realm ToTeams OracleSBC1 (local-policy-attribute)# done OracleSBC1 (local-policy)# done |
| Local policy From ANY to Teams (REFER Method) | OracleSBC1# configure terminal OracleSBC1 (configure)# session-router OracleSBC1 (session-router)# local-policy OracleSBC1 (local-policy)# from-address * OracleSBC1 (local-policy)# to-address * OracleSBC1 (local-policy)# source-realm * OracleSBC1 (local-policy)# policy-attribute OracleSBC1 (policy-attribute)# methods REFER OracleSBC1 (policy-attribute)# next-hop sag:TeamsGrp OracleSBC1 (policy-attribute)# realm ToTeams OracleSBC1 (local-policy-attribute)# done OracleSBC1 (local-policy)# done |

➤ Codec policy

| Element | Configuration |
|---------------------|---|
| Codec policy | OracleSBC1# configure terminal OracleSBC1 (configure)# media-manager OracleSBC1 (media-manager)# codec-policy OracleSBC1 (codec-policy)# name CodecToTeams OracleSBC1 (codec-policy)# allow-codecs (G722 PCMA PCMU G729 telephone-event) |

| | |
|--|--|
| | OracleSBC1 (codec-policy)# order-codecs (G722 PCMA PCMU G729) OracleSBC1 (codec-policy)# done |
|--|--|

➤ Media policy

Signaling DSCP marking 24 (CS3)

| Element | Configuration |
|---------------------|---|
| Codec policy | OracleSBC1# configure terminal OracleSBC1 (configure)# media-manager OracleSBC1 (media-manager)# media-policy OracleSBC1 (media-policy)# name DSCP OracleSBC1 (media-policy)# tos-settings OracleSBC1 (tos-settings)# media-type message OracleSBC1 (tos-settings)# media-sub-type sip OracleSBC1 (tos-settings)# tos-value 0x60 OracleSBC1 (tos-settings)# done |

Audio DSCP marking 46 (EF)

| Element | Configuration |
|---------------------|--|
| Codec policy | OracleSBC1# configure terminal OracleSBC1 (configure)# media-manager OracleSBC1 (media-manager)# media-policy OracleSBC1 (media-policy)# tos-settings OracleSBC1 (tos-settings)# media-type audio OracleSBC1 (tos-settings)# tos-value 0xb8 OracleSBC1 (tos-settings)# done |

➤ Sip manipulation

SIP Manipulation FixCLineNAT:

The following manipulation will change SDP "c" line to change private IP address to public IP address when the SBC is behind NAT device.

| Element | Configuration |
|---|--|
| SIP Manipulation FixCLineNAT | OracleSBC1# configure terminal OracleSBC1 (configure)# session-router OracleSBC1 (session-router)# sip-manipulation OracleSBC1 (sip-manipulation)# name FixCLineNATOutbound |
| SIP Mime SDP rule MSR_FixCLine | OracleSBC1 (sip-manipulation)# mime-sdp-rules OracleSBC1 (sip-mime-sdp-rules)# name MSR_FixCLine OracleSBC1 (sip-mime-sdp-rules)# msg-type any OracleSBC1 (sip-mime-sdp-rules)# comparison-type case-sensitive OracleSBC1 (sip-mime-sdp-rules)# action manipulate OracleSBC1 (sip-mime-sdp-rules)# sdp-session-rule |

| | |
|---|--|
| | <pre> OracleSBC1 (sip-sdp-session-rules)# name SR_FixCLine OracleSBC1 (sip-sdp-session-rules)# action manipulate OracleSBC1 (sip-mime-sdp-rules)# comparison-type case-sensitive OracleSBC1 (sip-sdp-session-rules)# sdp-line-rules OracleSBC1 (sip-sdp-line-rules)# name LR_FixCLine OracleSBC1 (sip-sdp-line-rules)# type c OracleSBC1 (sip-sdp-line-rules)# action find-replace-all OracleSBC1 (sip-sdp-line-rules)# match-value IN IP4 <Private @IP> OracleSBC1 (sip-sdp-line-rules)# new-value "IN IP4 <SBC Public @IP>" OracleSBC1 (sip-sdp-line-rules)# done OracleSBC1 (sip-sdp-line-rules)# exit OracleSBC1 (sip-sdp-session-rules)# done OracleSBC1 (sip-sdp-session-rules)# exit OracleSBC1 (sip-mime-sdp-rules)# exit OracleSBC1 (sip-mime-sdp-rules)# done </pre> |
| <p>SIP Header rule HR_FixCLine</p> | <pre> OracleSBC1 (sip-manipulation)# header-rules OracleSBC1 (sip-header-rules)# name HR_FixCLine OracleSBC1 (sip-header-rules)# header-name Content-Type OracleSBC1 (sip-header-rules)# action manipulation OracleSBC1 (sip-header-rules)# element-rule OracleSBC1 (sip-element-rules)# name ER_FixCLine OracleSBC1 (sip-element-rules)# parameter-name application/sdp OracleSBC1 (sip-element-rules)# type mime OracleSBC1 (sip-element-rules)# action find-replace-all OracleSBC1 (sip-element-rules)# match-value <Private @IP> OracleSBC1 (sip-element-rules)# new-value <Public @IP> OracleSBC1 (sip-element-rules)# done OracleSBC1 (sip-element-rules)# exit OracleSBC1 (sip-header-rules)# done OracleSBC1 (sip-header-rules)# exit OracleSBC1 (sip-manipulation)# done </pre> |

SIP Manipulation KeepFirstCodec

| Element | Configuration |
|---|---|
| <p>SIP Manipulation KeepFirstCodec</p> | <pre> OracleSBC1# configure terminal OracleSBC1 (configure)# session-router OracleSBC1 (session-router)# sip-manipulation OracleSBC1 (sip-manipulation)# name KeepFirstCodec </pre> |
| <p>SIP Mime SDP rule</p> | <pre> OracleSBC1 (sip-manipulation)# mime-sdp-rules OracleSBC1 (sip-mime-sdp-rules)# name MSR_KeepFirstCodec OracleSBC1 (sip-mime-sdp-rules)# msg-type any OracleSBC1 (sip-mime-sdp-rules)# action manipulate OracleSBC1 (sip-mime-sdp-rules)# sdp-media-rule OracleSBC1 (sip-sdp-media-rules)# name MR_keepFirstCodec OracleSBC1 (sip-sdp-media-rules)# media-type audio[0] OracleSBC1 (sip-sdp-media-rules)# action manipulate OracleSBC1 (sip-sdp-media-rules)# comparison-type case-sensitive OracleSBC1 (sip-sdp-media-rules)# sdp-line-rules OracleSBC1 (sip-sdp-line-rules)# name LR_firstCodec OracleSBC1 (sip-sdp-line-rules)# type m OracleSBC1 (sip-sdp-line-rules)# action store OracleSBC1 (sip-sdp-line-rules)# comparison-type pattern-rule </pre> |

| | |
|---------------------------|---|
| MSR_KeepFirstCodec | <pre> OracleSBC1 (sip-sdp-line-rules)# match-value ^(*RTP/SAVP)([0-9]{1,3})(.*)\$ OracleSBC1 (sip-sdp-line-rules)# done OracleSBC1 (sip-sdp-line-rules)# exit OracleSBC1 (sip-sdp-media-rules)# sdp-line-rule OracleSBC1 (sip-sdp-line-rules)# name LR_mLineFirstCodecOnly OracleSBC1 (sip-sdp-line-rules)# type m OracleSBC1 (sip-sdp-line-rules)# action replace OracleSBC1 (sip-sdp-line-rules)# comparison-type boolean OracleSBC1 (sip-sdp-line-rules)# match-value \$MSR_keepFirstCodec.\$MR_keepFirstCodec.\$LR_firstCodec OracleSBC1 (sip-sdp-line-rules)# new-value \$MSR_keepFirstCodec.\$MR_keepFirstCodec.\$LR_firstCodec.\$1+\$MSR_ keepFirstCodec.\$MR_keepFirstCodec.\$LR_firstCodec.\$2+ " 101" OracleSBC1 (sip-sdp-line-rules)# done OracleSBC1 (sip-sdp-line-rules)# exit OracleSBC1 (sip-sdp-media-rules)# sdp-line-rule OracleSBC1 (sip-sdp-line-rules)# name LR_aLineFirstCodecOnly OracleSBC1 (sip-sdp-line-rules)# type a OracleSBC1 (sip-sdp-line-rules)# action delete OracleSBC1 (sip-sdp-line-rules)# comparison-type pattern-rule OracleSBC1 (sip-sdp-line-rules)# match-value (rtpmap fmt):(?!((\$MSR_keepFirstCodec.\$MR_keepFirstCodec.\$LR_ firstCodec.\$2} 101)) OracleSBC1 (sip-sdp-line-rules)# done OracleSBC1 (sip-sdp-line-rules)# exit OracleSBC1 (sip-mime-sdp-rules)# done OracleSBC1 (sip-mime-sdp-rules)# exit </pre> |
|---------------------------|---|

SIP Manipulation AddAnnexBNo

| Element | Configuration |
|--|--|
| SIP Manipulation AddAnnexBNo | <pre> OracleSBC1# configure terminal OracleSBC1 (configure)# session-router OracleSBC1 (session-router)# sip-manipulation OracleSBC1 (sip-manipulation)# name AddAnnexBNo </pre> |
| SIP Mime SDP rule MSR_AddAnnexBNo | <pre> OracleSBC1 (sip-manipulation)# mine-sdp-rules OracleSBC1 (sip-mime-sdp-rules)# name MSR_AddAnnexBNo OracleSBC1 (sip-mime-sdp-rules)# msg-type any OracleSBC1 (sip-mime-sdp-rules)# methods INVITE OracleSBC1 (sip-mime-sdp-rules)# action manipulate OracleSBC1 (sip-mime-sdp-rules)# sdp-media-rule OracleSBC1 (sip-sdp-media-rules)# name MR_TestG729 OracleSBC1 (sip-sdp-media-rules)# media-type audio OracleSBC1 (sip-sdp-media-rules)# action manipulate OracleSBC1 (sip-sdp-media-rules)# comparison-type case-sensitive OracleSBC1 (sip-sdp-media-rules)# sdp-line-rules OracleSBC1 (sip-sdp-line-rules)# name LR_MatchG729 OracleSBC1 (sip-sdp-line-rules)# type m OracleSBC1 (sip-sdp-line-rules)# action store OracleSBC1 (sip-sdp-line-rules)# comparison-type pattern-rule OracleSBC1 (sip-sdp-line-rules)# match-value \s18\b OracleSBC1 (sip-sdp-line-rules)# done </pre> |



| | |
|--|--|
| | <pre> OracleSBC1 (sip-sdp-line-rules)# exit OracleSBC1 (sip-sdp-media-rules)# sdp-line-rule OracleSBC1 (sip-sdp-line-rules)# name LR_Delete OracleSBC1 (sip-sdp-line-rules)# type a OracleSBC1 (sip-sdp-line-rules)# action delete OracleSBC1 (sip-sdp-line-rules)# comparison-type pattern-rule OracleSBC1 (sip-sdp-line-rules)# match-value ^.*annexb.*\$ OracleSBC1 (sip-sdp-line-rules)# done OracleSBC1 (sip-sdp-line-rules)# exit OracleSBC1 (sip-sdp-media-rules)# name MR_AddAnnexBno0 OracleSBC1 (sip-sdp-media-rules)# media-type audio[0] OracleSBC1 (sip-sdp-media-rules)# action manipulate OracleSBC1 (sip-sdp-media-rules)# comparison-type boolean OracleSBC1 (sip-sdp-media-rules)# match-value \$MSR_AddAnnexBNo.\$ MR_TestG729.\$LR_MatchG729[0] OracleSBC1 (sip-sdp-media-rules)# sdp-line-rule OracleSBC1 (sip-sdp-line-rules)# name LR_AddAnnexbNo OracleSBC1 (sip-sdp-line-rules)# type a OracleSBC1 (sip-sdp-line-rules)# action add OracleSBC1 (sip-sdp-line-rules)# new-value "fmtp:18 annexb=no" OracleSBC1 (sip-sdp-line-rules)# done OracleSBC1 (sip-sdp-line-rules)# exit OracleSBC1 (sip-mime-sdp-rules)# done OracleSBC1 (sip-mime-sdp-rules)# exit </pre> |
|--|--|

SIP Manipulation DelReasonCause34:

| Element | Configuration |
|--|---|
| SIP Manipulation DelReasonCause34 | <pre> OracleSBC1# configure terminal OracleSBC1 (configure)# session-router OracleSBC1 (session-router)# sip-manipulation OracleSBC1 (sip-manipulation)# name DelReasonHeaderCause34 </pre> |
| SIP Header rule HR_ DelReasonHeader | <pre> OracleSBC1 (sip-manipulation)# header-rules OracleSBC1 (sip-header-rule)# name HR_is486 OracleSBC1 (sip-header-rule)# header-name @status-line OracleSBC1 (sip-header-rule)# action store OracleSBC1 (sip-header-rule)# comparison-type pattern-rule OracleSBC1 (sip-header-rule)# msg-type reply OracleSBC1 (sip-header-rule)# element-type OracleSBC1 (sip-element-rule)# name is486Code OracleSBC1 (sip-element-rule)# type status-code OracleSBC1 (sip-element-rule)# action store OracleSBC1 (sip-element-rule)# match-val-type any OracleSBC1 (sip-element-rule)# comparison-type pattern-rule OracleSBC1 (sip-element-rule)# match-value 486 OracleSBC1 (sip-element-rule)# done OracleSBC1 (sip-element-rule)# exit OracleSBC1 (sip-header-rule)# done OracleSBC1 (sip-header-rule)# name HR_isCause34 OracleSBC1 (sip-header-rule)# header-name Reason OracleSBC1 (sip-header-rule)# action store OracleSBC1 (sip-header-rule)# comparison-type pattern-rule OracleSBC1 (sip-header-rule)# msg-type reply </pre> |



| | |
|--|--|
| | <pre> OracleSBC1 (sip-header-rule)# match-value ^.*cause=34.*\$ OracleSBC1 (sip-header-rule)# done OracleSBC1 (sip-header-rule)# name HR_DelReasonHeader OracleSBC1 (sip-header-rule)# header-name Reason OracleSBC1 (sip-header-rule)# action delete OracleSBC1 (sip-header-rule)# comparison-type pattern-rule OracleSBC1 (sip-header-rule)# msg-type reply OracleSBC1 (sip-header-rule)# match-value "\$HR_is486.\$is486Code & \$HR_isCause34" OracleSBC1 (sip-header-rule)# done OracleSBC1 (sip-header-rule)# exit OracleSBC1 (sip-manipulation)# done </pre> |
|--|--|

SIP Manipulation RemovePrivacy:

| Element | Configuration |
|---|---|
| SIP Manipulation RemovePrivacy | <pre> OracleSBC1# configure terminal OracleSBC1 (configure)# session-router OracleSBC1 (session-router)# sip-manipulation OracleSBC1 (sip-manipulation)# name RemovePrivacy </pre> |
| SIP Header rule HR_RemovePrivacy | <pre> OracleSBC1 (sip-manipulation)# header-rules OracleSBC1 (sip-header-rule)# name HR_RemovePrivacy OracleSBC1 (sip-header-rule)# header-name Privacy OracleSBC1 (sip-header-rule)# action delete OracleSBC1 (sip-header-rule)# comparison-type boolean OracleSBC1 (sip-header-rule)# msg-type out-of-dialog OracleSBC1 (sip-header-rule)# methods INVITE,BYE,REFER OracleSBC1 (sip-header-rule)# match-value \$FROM_USER.\$0!=anonymous OracleSBC1 (sip-header-rule)# done OracleSBC1 (sip-header-rule)# exit OracleSBC1 (sip-manipulation)# done </pre> |

SIP Manipulation FixALineNATInbound:

| Element | Configuration |
|--|---|
| SIP Manipulation FixALineNATInbound | <pre> OracleSBC# configure terminal OracleSBC (configure)# session-router OracleSBC (session-router)# sip-manipulation OracleSBC (sip-manipulation)# name FixALineNATInbound </pre> |
| Header rule HR_FixALineNATInbound | <pre> OracleSBC (sip-manipulation)# header-rules OracleSBC (sip-header-rules)# name HR_FixALineNATInbound OracleSBC (sip-header-rules)# header-name Content-Type OracleSBC (sip-header-rules)# action manipulate OracleSBC (sip-header-rules)# msg-type any OracleSBC (sip-header-rules)# element-rule OracleSBC (sip-element-rules)# name ER_FixALineNATInbound OracleSBC (sip-element-rules)# parameter-name application/sdp OracleSBC (sip-element-rules)# type mime OracleSBC (sip-element-rules)# action find-replace-all </pre> |

| | |
|--|---|
| | <pre> OracleSBC (sip-element-rules)# match-value <SBC Public @IP> OracleSBC (sip-element-rules)# new-value <SBC Private @IP> OracleSBC (sip-element-rules)# done OracleSBC (sip-element-rules)# exit OracleSBC (sip-header-rules)# done OracleSBC (sip-header-rules)# exit OracleSBC (sip-manipulation)# done </pre> |
|--|---|

SIP Manipulation ChangeUserAgent:

| Element | Configuration |
|--|---|
| <p>SIP Manipulation</p> <p>ChangeUserAgent</p> | <pre> OracleSBC1# configure terminal OracleSBC1 (configure)# session-router OracleSBC1 (session-router)# sip-manipulation OracleSBC1 (sip-manipulation)# name ChangeUserAgent </pre> |
| <p>Header rule</p> <p>HR_ChangeUserAgent</p> | <pre> OracleSBC1 (sip-manipulation)# header-rules OracleSBC1 (sip-header-rules)# name HR_ChangeUserAgent OracleSBC1 (sip-header-rules)# header-name User-Agent OracleSBC1 (sip-header-rules)# action manipulate OracleSBC1 (sip-header-rules)# msg-type request OracleSBC1 (sip-header-rules)# methods INVITE OracleSBC1 (sip-header-rules)# new-value "ORACLE SBC/v.8.4.0 \\ Teams" OracleSBC1 (sip-header-rules)# done OracleSBC1 (sip-header-rules)# exit OracleSBC1 (sip-manipulation)# done </pre> |

SIP Manipulation ChangeServer:

| Element | Configuration |
|---|---|
| <p>SIP manipulation</p> <p>ChangeServer</p> | <pre> OracleSBC1# configure terminal OracleSBC1 (configure)# session-router OracleSBC1 (session-router)# sip-manipulation OracleSBC1 (sip-manipulation)# name ChangeServer </pre> |
| <p>Header rule</p> <p>ChangeServer</p> | <pre> OracleSBC1 (sip-manipulation)# header-rules OracleSBC1 (sip-header-rules)# name HR_ChangeServer OracleSBC1 (sip-header-rules)# header-name Server OracleSBC1 (sip-header-rules)# action manipulate OracleSBC1 (sip-header-rules)# msg-type reply OracleSBC1 (sip-header-rules)# new-value "ORACLE SBC/v.8.4.0 \\ Teams" OracleSBC1 (sip-header-rules)# done OracleSBC1 (sip-header-rules)# exit OracleSBC1 (sip-manipulation)# done </pre> |

SIP Manipulation StripVideo

| Element | Configuration |
|-------------------------------------|--|
| SIP Manipulation StripVideo | OracleSBC1# configure terminal OracleSBC1 (configure)# session-router OracleSBC1 (session-router)# sip-manipulation OracleSBC1 (sip-manipulation)# name StripVideo |
| SIP Mime SDP rule MSR_StripVideo | OracleSBC1 (sip-manipulation)# mime-sdp-rules OracleSBC1 (sip-mime-sdp-rules)# name MSR_StripVideo OracleSBC1 (sip-mime-sdp-rules)# msg-type request OracleSBC1 (sip-mime-sdp-rules)# methods INVITE OracleSBC1 (sip-mime-sdp-rules)# comparison-type manipulate OracleSBC1 (sip-mime-sdp-rules)# sdp-media-rule OracleSBC1 (sip-sdp-media-rules)# name MR_RemoveVideo OracleSBC1 (sip-sdp-media-rules)# media-type video OracleSBC1 (sip-sdp-media-rules)# action delete OracleSBC1 (sip-sdp-media-rules)# done OracleSBC1 (sip-sdp-media-rules)# exit OracleSBC1 (sip-mime-sdp-rules)# done OracleSBC1 (sip-mime-sdp-rules)# exit |

SIP Manipulation out-teams:

| Element | Configuration |
|---------------------------------------|---|
| SIP manipulation out-teams | OracleSBC1# configure terminal OracleSBC1 (configure)# session-router OracleSBC1 (session-router)# sip-manipulation OracleSBC1 (sip-manipulation)# name out-teams |
| SIP Header rule HR_CallFixCLineNAT | OracleSBC1 (sip-manipulation)# header-rules OracleSBC1 (sip-header-rules)# name CallFixCLineNATOutbound OracleSBC1 (sip-header-rules)# header-name From OracleSBC1 (sip-header-rules)# action sip-manip OracleSBC1 (sip-header-rules)# new-value FixCLineNATOutbound OracleSBC1 (sip-header-rules)# done OracleSBC1 (sip-header-rules)# exit OracleSBC1 (sip-manipulation)# done |
| SIP Header rule KeepFirstCodec | OracleSBC1 (sip-manipulation)# header-rules OracleSBC1 (sip-header-rules)# name CallKeepFirstCodec OracleSBC1 (sip-header-rules)# header-name From OracleSBC1 (sip-header-rules)# action sip-manip OracleSBC1 (sip-header-rules)# new-value KeepFirstCodec OracleSBC1 (sip-header-rules)# done OracleSBC1 (sip-header-rules)# exit OracleSBC1 (sip-manipulation)# done |

SIP Manipulation in-teams:

| Element | Configuration |
|---------|---------------|
|---------|---------------|



| | |
|---|--|
| SIP manipulation in-teams | OracleSBC1# configure terminal OracleSBC1 (configure)# session-router OracleSBC1 (session-router)# sip-manipulation OracleSBC1 (sip-manipulation)# name in-teams |
| SIP Header rule CallRemovePrivacy | OracleSBC1 (sip-manipulation)# header-rules OracleSBC1 (sip-header-rules)# name CallRemovePrivacy OracleSBC1 (sip-header-rules)# header-name Privacy OracleSBC1 (sip-header-rules)# action sip-manip OracleSBC1 (sip-header-rules)# new-value RemovePrivacy OracleSBC1 (sip-header-rules)# done |
| SIP Header rule CallFixALineNATInbound | OracleSBC (sip-header-rules)# name CallFixALineNATInbound OracleSBC (sip-header-rules)# header-name From OracleSBC (sip-header-rules)# action sip-manip OracleSBC (sip-header-rules)# new-value FixALineNATInbound OracleSBC (sip-header-rules)# done |
| SIP Header rule CallAddAnnexBNo | OracleSBC (sip-header-rules)# name CallAddAnnexBNo OracleSBC (sip-header-rules)# header-name From OracleSBC (sip-header-rules)# action sip-manip OracleSBC (sip-header-rules)# new-value AddAnnexBNo OracleSBC (sip-header-rules)# done OracleSBC (sip-header-rules)# exit OracleSBC (sip-manipulation)# done |
| SIP Header rule CallDelReasonHeaderCause34 | OracleSBC (sip-header-rules)# name CallDelReasonHeaderCause34 OracleSBC (sip-header-rules)# header-name From OracleSBC (sip-header-rules)# action sip-manip OracleSBC (sip-header-rules)# new-value DelReasonHeaderCause34 OracleSBC (sip-header-rules)# done OracleSBC (sip-header-rules)# exit OracleSBC (sip-manipulation)# done |
| SIP Header rule CallStripVideo | OracleSBC (sip-header-rules)# name CallStripVideo OracleSBC (sip-header-rules)# header-name From OracleSBC (sip-header-rules)# action sip-manip OracleSBC (sip-header-rules)# new-value StripVideo OracleSBC (sip-header-rules)# done OracleSBC (sip-header-rules)# exit OracleSBC (sip-manipulation)# done |

SIP Manipulation out-btip:

| Element | Configuration |
|--|---|
| SIP manipulation out-btip | OracleSBC1# configure terminal OracleSBC1 (configure)# session-router OracleSBC1 (session-router)# sip-manipulation OracleSBC1 (sip-manipulation)# name out-btip |
| SIP Header rule CallChangeUserAgent | OracleSBC1 (sip-manipulation)# header-rules OracleSBC1 (sip-header-rules)# name CallChangeUserAgent OracleSBC1 (sip-header-rules)# header-name User-Agent OracleSBC1 (sip-header-rules)# action sip-manip OracleSBC1 (sip-header-rules)# new-value ChangeUserAgent |

| | |
|---|---|
| | OracleSBC1 (sip-header-rules)# done |
| SIP Header rule CallChangeServer | OracleSBC1 (sip-header-rules)# name CallChangeServer OracleSBC1 (sip-header-rules)# header-name From OracleSBC1 (sip-header-rules)# action sip-manip OracleSBC1 (sip-header-rules)# new-value ChangeServer OracleSBC1 (sip-header-rules)# done |

➤ **Consultative transfer**

| Element | Configuration |
|------------------------------|---|
| Consultative transfer | OracleSBC1# configure terminal OracleSBC1 (configure)# session-router OracleSBC1 (session-router)# sip-feature OracleSBC1 (sip-feature)# name replaces OracleSBC1 (sip-feature)# realm ToTeams OracleSBC1 (sip-feature)# support-mode-inbound Pass OracleSBC1 (sip-feature)# require-mode-inbound Pass OracleSBC1 (sip-feature)# proxy-require-mode-inbound Pass OracleSBC1 (sip-feature)# support-mode-outbound Pass OracleSBC1 (sip-feature)# require-mode-outbound Pass OracleSBC1 (sip-feature)# proxy-require-mode-outbound Pass OracleSBC1 (sip-feature)# done |

➤ **SIP Profile**

| Element | Configuration |
|--------------------|--|
| SIP profile | OracleSBC1# configure terminal OracleSBC1 (configure)# session-router OracleSBC1 (session-router)# sip-profile OracleSBC1 (sip-profile)# name foreplaceTeams OracleSBC1 (sip-profile)# redirection inherit OracleSBC1 (sip-profile)# ingress-conditional-cac-admit inherit OracleSBC1 (sip-profile)# egress-conditional-cac-admit inherit OracleSBC1 (sip-profile)# forked-cac-bw inherit OracleSBC1 (sip-profile)# cnam-lookup-dir egress OracleSBC1 (sip-profile)# replace-dialogs enabled OracleSBC1 (sip-profile)# done |

➤ **SDES profile**

| Element | Configuration |
|---------------------|---|
| SDES profile | OracleSBC1# configure terminal OracleSBC1 (configure)# security OracleSBC1 (security)# media-security OracleSBC1 (media-security)# sdes-profile OracleSBC1 (sdes-profile)# name SDES OracleSBC1 (sdes-profile)# crypto-list AES_CM_128_HMAC_SHA1_80 OracleSBC1 (sdes-profile)# srtplib-auth enabled OracleSBC1 (sdes-profile)# srtplib-encrypt enabled |

| | |
|--|---|
| | OracleSBC1 (sdes-profile)# srtcp-encrypt enabled OracleSBC1 (sdes-profile)# mki disabled OracleSBC1 (sdes-profile)# egress-offer-format same-as-ingress OracleSBC1 (sdes-profile)# srtcp-rekey-on-re-invite disabled OracleSBC1 (sdes-profile)# done |
|--|---|

➤ **RTCP policy**

| Element | Configuration |
|--------------------|---|
| RTCP policy | OracleSBC1# configure terminal OracleSBC1 (configure)# media-manager OracleSBC1 (media-manager)# rtcp-policy OracleSBC1 (rtcp-policy)# name rtcpGen OracleSBC1 (rtcp-policy)# rtcp-generate all-calls OracleSBC1 (rtcp-policy)# hide-cname disabled OracleSBC1 (rtcp-policy)# done |

➤ **Net Management Control**

| Element | Configuration |
|-------------------------------|--|
| Net-Management-Control | OracleSBC1# configure terminal OracleSBC1 (configure)# session-router OracleSBC1 (session-router)# net-management-control OracleSBC1 (net-management-control)# name Emergency OracleSBC1 (net-management-control)# type priority OracleSBC1 (net-management-control)# treatment apply-local-policy OracleSBC1 (net-management-control)# protocol-next-hop SIP OracleSBC1 (net-management-control)# destination-identifier (+33112 +3318 +3317 +3315 +33114) OracleSBC1 (net-management-control)# done |

➤ **Session constraint**

| Element | Configuration |
|---------------------------|---|
| Session-constraint | OracleSBC1# configure terminal OracleSBC1 (configure)# session-router OracleSBC1 (session-router)# session-constraints OracleSBC1 (session-constraints)# name SessionConstraintEmergency OracleSBC1 (session-constraints)# max-sessions <X> e.g. 50 OracleSBC1 (session-constraints)# done |

➤ Media sec policy

| Element | Configuration |
|------------------|---|
| Media sec policy | OracleSBC1# configure terminal |
| | OracleSBC1 (configure)# security |
| | OracleSBC1 (security)# media-security |
| | OracleSBC1 (media-security)# media-sec-policy |
| | OracleSBC1 (media-sec-policy)# name RTP |
| | OracleSBC1 (media-sec-policy)# pass-through disabled |
| | OracleSBC1 (media-sec-policy)# inbound |
| | OracleSBC1 (inbound)# mode rtp |
| | OracleSBC1 (inbound)# protocol none |
| | OracleSBC1 (inbound)# hide-egress-media-update enabled |
| | OracleSBC1 (inbound)# done |
| | OracleSBC1 (inbound)# exit |
| | OracleSBC1 (media-sec-policy)# outbound |
| | OracleSBC1 (outbound)# mode rtp |
| | OracleSBC1 (outbound)# protocol none |
| | OracleSBC1 (outbound)# done |
| | OracleSBC1 (outbound)# exit |
| | OracleSBC1 (media-sec-policy)# name SRTP |
| | OracleSBC1 (media-sec-policy)# pass-through disabled |
| | OracleSBC1 (media-sec-policy)# inbound |
| | OracleSBC1 (inbound)# profile SDES |
| | OracleSBC1 (inbound)# mode srtp |
| | OracleSBC1 (inbound)# protocol sdes |
| | OracleSBC1 (inbound)# done |
| | OracleSBC1 (inbound)# exit |
| | OracleSBC1 (media-sec-policy)# outbound |
| | OracleSBC1 (outbound)# profile SDES |
| | OracleSBC1 (outbound)# mode srtp |
| | OracleSBC1 (outbound)# protocol sdes |
| | OracleSBC1 (outbound)# done |

➤ ICE-Profile

| Element | Configuration |
|-------------|--|
| ICE-Profile | OracleSBC# configure terminal |
| | OracleSBC (configure)# media-manager |
| | OracleSBC (media-manager)# ice-profile |
| | OracleSBC (ice-profile)# name ice-profile |
| | OracleSBC (ice-profile)# stun-conn-timeout 0 |
| | OracleSBC (ice-profile)# stun-keep-alive-interval 0 |
| | OracleSBC (ice-profile)# stun-rate-limit 100 |
| | OracleSBC (ice-profile)# done |

➤ Teams Realm

| Element | Configuration |
|---------|---------------------------------------|
| Realm | OracleSBC1# configure terminal |

| | |
|---------|---|
| ToTeams | <pre> OracleSBC1 (configure)# media-manager OracleSBC1 (media-manager)# realm-config OracleSBC1 (realm-config)# identifier ToTeams OracleSBC1 (realm-config)# network-interfaces s0p0:0 OracleSBC1 (realm-config)# mm-in-realm enabled OracleSBC1 (realm-config)# qos-enable enabled OracleSBC1 (realm-config)# net-management-control enabled OracleSBC1 (realm-config)# refer-call-transfer enabled OracleSBC1 (realm-config)# media-sec-policy SRTP OracleSBC1 (realm-config)# rtcp-policy rtcpGen OracleSBC1 (realm-config)# rtcp-mux enabled OracleSBC1 (realm-config)# codec-policy CodecToTeams OracleSBC1 (realm-config)# ice-profile ice-profile OracleSBC1 (realm-config)# media-realm-list ToTeams OracleSBC1 (realm-config)# teams-fqdn <SBC FQDN> OracleSBC1 (realm-config)# teams-fqdn-in-uri enabled OracleSBC1 (realm-config)# sdp-inactive-only enabled OracleSBC1 (realm-config)# done </pre> |
|---------|---|

➤ SIP interface

| Element | Configuration |
|------------------------------|---|
| SIP interface ToTeams | <pre> OracleSBC1# configure terminal OracleSBC1 (configure)# session-router OracleSBC1 (session-router)# sip-interface OracleSBC1 (sip-interface)# state enabled OracleSBC1 (sip-interface)# realm-id ToTeams OracleSBC1 (sip-interface)# in-manipulationid in-teams OracleSBC1 (sip-interface)# out-manipulationid out-teams OracleSBC1 (sip-interface)# sip-profile foreplaceTeams OracleSBC1 (sip-interface)# add-sdp-invite reinvoke OracleSBC1 (sip-interface)# sip-port OracleSBC1 (sip-port)# address <SBC Public IP> OracleSBC1 (sip-port)# port 5061 OracleSBC1 (sip-port)# transport-protocol TLS OracleSBC1 (sip-port)# tls-profile TeamsTLS OracleSBC1 (sip-port)# allow-anonymous agents-only OracleSBC1 (sip-port)# done </pre> |

6.3.4 BTIP configuration

➤ Enable SIP config

| Element | Configuration |
|------------|--|
| SIP config | <pre> OracleSBC1# configure terminal OracleSBC1 (configure)# session-router OracleSBC1# (session-router)# sip-config OracleSBC1 (sip-config)# home-realm-id ToBTIP OracleSBC1 (sip-config)# registrar-domain * OracleSBC1 (sip-config)# registrar-host * OracleSBC1 (sip-config)# registrar-port 5060 OracleSBC1 (sip-config)# dialog-transparency disabled OracleSBC1 (sip-config)# options +inmanip-before-validate OracleSBC1 (sip-config)# options +max-udp-length=0 OracleSBC1 (sip-config)# options +reinvite-trying=yes OracleSBC1 (sip-config)# options +multiple-dialogs-enhancement </pre> |

➤ SIP Profile

| Element | Configuration |
|-------------|---|
| SIP profile | <pre> OracleSBC1# configure terminal OracleSBC1 (configure)# session-router OracleSBC1 (session-router)# sip-profile OracleSBC1 (sip-profile)# name foreplaceBTIP OracleSBC1 (sip-profile)# redirection inherit OracleSBC1 (sip-profile)# ingress-conditional-cac-admit inherit OracleSBC1 (sip-profile)# egress-conditional-cac-admit inherit OracleSBC1 (sip-profile)# forked-cac-bw inherit OracleSBC1 (sip-profile)# cnam-lookup-dir egress OracleSBC1 (sip-profile)# replace-dialogs disabled OracleSBC1 (sip-profile)# done </pre> |

➤ Codec policy

| Element | Configuration |
|--------------------------|---|
| Codec policy : CodecBTIP | <pre> OracleSBC1# configure terminal OracleSBC1 (configure)# media-manager OracleSBC1 (media-manager)# codec-policy OracleSBC1 (codec-policy)# name CodecBTIP OracleSBC1 (codec-policy)# allow-codecs (G722 PCMA G729 telephone-event) OracleSBC1 (codec-policy)# done </pre> |

| Element | Configuration |
|------------------------|---|
| Codec policy : CodecBT | <pre> OracleSBC1# configure terminal OracleSBC1 (configure)# media-manager OracleSBC1 (media-manager)# codec-policy OracleSBC1 (codec-policy)# name CodecBT OracleSBC1 (codec-policy)# allow-codex (G722 PCMA PCMU G729 telephone-event) OracleSBC1 (codec-policy)# done </pre> |

➤ Session agent

| Element | Configuration |
|-------------------------|--|
| Session agent active | <pre> OracleSBC1# configure terminal OracleSBC1 (configure)# session-router OracleSBC1 (session-router)# session-agent OracleSBC1 (session-agent)# hostname <@IP aSBC 1> OracleSBC1 (session-agent)# port 5060 OracleSBC1 (session-agent)# transport-method StaticTCP OracleSBC1 (session-agent)# realm-id ToBTIP OracleSBC1 (session-agent)# ping-response enabled OracleSBC1 (session-agent)# ping-method OPTIONS OracleSBC1 (session-agent)# ping-interval 300 OracleSBC1 (session-agent)# refer-call-transfer enabled If session agent = BTIP (France). Then : OracleSBC1 (session-agent)# codec-policy CodecBTIP Else If session agent = BT(Int). Then : OracleSBC1 (session-agent)# codec-policy CodecBT OracleSBC1 (session-agent)# done </pre> |
| Session agent backup | <pre> OracleSBC1# configure terminal OracleSBC1 (configure)# session-router OracleSBC1 (session-router)# session-agent OracleSBC1 (session-agent)# hostname <@IP aSBC 2> OracleSBC1 (session-agent)# port 5060 OracleSBC1 (session-agent)# transport-method StaticTCP OracleSBC1 (session-agent)# realm-id ToBTIP OracleSBC1 (session-agent)# ping-response enabled OracleSBC1 (session-agent)# ping-method OPTIONS OracleSBC1 (session-agent)# ping-interval 300 OracleSBC1 (session-agent)# refer-call-transfer enabled If session agent = BTIP (France). Then : OracleSBC1 (session-agent)# codec-policy CodecBTIP Else If session agent = BT(Int). Then : OracleSBC1 (session-agent)# codec-policy CodecBT OracleSBC1 (session-agent)# done </pre> |

➤ Session group

| Element | Configuration |
|---------------|--|
| Session group | <pre> OracleSBC1# configure terminal OracleSBC1 (configure)# session-router </pre> |

| | |
|--|--|
| | <pre> OracleSBC1 (session-router)# session-group OracleSBC1 (session-group)# group-name BTIPGrp OracleSBC1 (session-group)# dest (<@IP aSBC 1> <@IP aSBC 2>) OracleSBC1 (session-group)# sag-recursion enable OracleSBC1 (session-group)# stop-sag-recurse 300-407,409-599 OracleSBC1 (session-group)# done </pre> |
|--|--|

➤ Realm

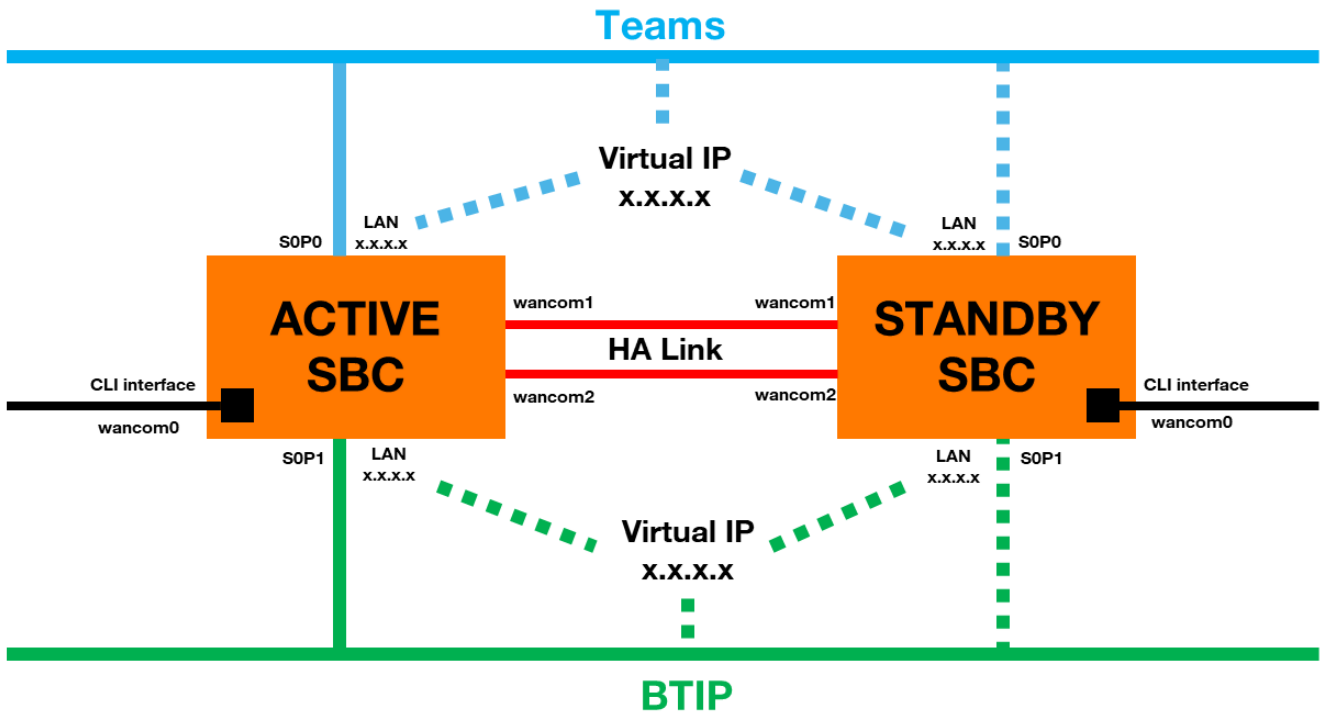
| Element | Configuration |
|--|--|
| <p>Realm</p> <p>ToBTIP</p> | <pre> OracleSBC1# configure terminal OracleSBC1 (configure)# media-manager OracleSBC1 (media-manager)# realm-config OracleSBC1 (realm-config)# identifier ToBTIP OracleSBC1 (realm-config)# network-interfaces s0p1:0 OracleSBC1 (realm-config)# mm-in-realm enabled OracleSBC1 (realm-config)# qos-enabled enabled OracleSBC1 (realm-config)# refer-call-transfer enabled OracleSBC1 (realm-config)# media-sec-policy RTP OracleSBC1 (realm-config)# codec-policy CodecBTIP OracleSBC1 (realm-config)# rtcp-policy rtcpGen OracleSBC1 (realm-config)# merge-early-dialog enable OracleSBC1 (realm-config)# hide-egress-media-update enabled OracleSBC1 (realm-config)# constraint-name SessionConstraintEmergency OracleSBC1 (realm-config)# ringback-trigger refer OracleSBC1 (realm-config)# ringback-file <ringback file .raw>* OracleSBC1 (realm-config)# done </pre> |

➤ SIP interface

| Element | Configuration |
|-----------------------------|--|
| <p>SIP interface</p> | <pre> OracleSBC1# configure terminal OracleSBC1 (configure)# session-router OracleSBC1 (session-router)# sip-interface OracleSBC1 (sip-interface)# state enabled OracleSBC1 (sip-interface)# realm-id ToBTIP OracleSBC1 (sip-interface)# sip-profile foreplaceBTIP OracleSBC1 (sip-interface)# initial-inv-trans-expire 6 OracleSBC1 (sip-interface)# out-manipulationid out-btip OracleSBC1 (sip-interface)# sip-port OracleSBC1 (sip-port)# address <Private @IP> OracleSBC1 (sip-port)# port 5060 OracleSBC1 (sip-port)# transport-protocol TCP OracleSBC1 (sip-port)# allow-anonymous agents-only OracleSBC1 (sip-port)# done </pre> |

6.4 ORACLE SBC - HA configuration

This section describes the Oracle SBC configuration in HA mode (Two SBC in a high availability mode). The HA topology validated within VISIT Teams offer is displayed in the figure below:



Only the Active SBC is used, while the second SBC is considered as a backup device (standby).

Following steps have to be performed to ensure correct integration within VISIT Teams offer:

- First SBC configuration
 - NTP synchronization
 - Virtual MAC address
 - Primary & Secondary utility address
 - Wancom1 & Wancom2 physical configuration
 - Wancom1 & Wancom2 network configuration
 - Redundancy
- Secondary SBC configuration

6.4.1 First SBC configuration

➤ NTP synchronization

| Element | Configuration |
|----------|--|
| NTP sync | <pre>OracleSBC1# configure terminal OracleSBC1 (configure)# ntp-sync OracleSBC1 (ntp-config)# add-server <NTP @IP> OracleSBC1 (ntp-config)# done</pre> |

➤ Virtual MAC address

| Element | Configuration |
|---------------------|---|
| Virtual MAC address | <pre>OracleSBC1# show interfaces</pre> |

Identify "Ethernet address is 00:08:25:XX:YY:ZN", 00:08:25 refers to Acme Packet, XX:YY:Z refers to the specific SBC. N is a 0-f hexadecimal value available for Oracle SBC. To create a virtual MAC address replace the "N" value with unused hexadecimal values for Oracle SBC: 8,9,e or f. Example:

```
Ethernet address s0p0: 00:08:25:A2:45:BF
Virtual MAC 1: 00:08:25:A2:45:B8
Virtual MAC 2: 00:08:25:A2:45:B9
```

| Element | Configuration |
|---------------------|--|
| Virtual MAC address | <pre>OracleSBC1# configure terminal OracleSBC1 (configure)# system OracleSBC1 (system)# phy-interface OracleSBC1 (phy-interface)# select <name>: 1: s0p0 2: s0p1 Selection: 1 OracleSBC1 (phy-interface)# virtual-mac <Virtual MAC 1> OracleSBC1 (phy-interface)# done OracleSBC1 (phy-interface)# select <name>: 1: s0p0 2: s0p1 Selection: 2 OracleSBC1 (phy-interface)# virtual-mac <Virtual MAC 2> OracleSBC1 (phy-interface)# done</pre> |



➤ Primary & Secondary utility address

Primary and Secondary utility addresses have to be in the same subnet than Public @IP (to teams) and Private @IP (to BTIP):

| Element | Configuration |
|-------------------------------------|--|
| Primary & Secondary utility address | <pre> OracleSBC1# configure terminal OracleSBC1 (configure)# system OracleSBC1 (system)# network-interface OracleSBC1 (network-interface)# select <name>:<sub-port-id>: 1: s0p0:0 ip=Public @IP gw= GW @IP 2: s0p1:0 ip=Private @IP gw= GW @IP Selection: 1 OracleSBC1 (network-interface)# pri-utility-addr <@IP> OracleSBC1 (network-interface)# sec-utility-addr <@IP> OracleSBC1 (network-interface)# done OracleSBC1 (network-interface)# select <name>:<sub-port-id>: 1: s0p0:0 ip=Public @IP gw= GW @IP 2: s0p1:0 ip=Private @IP gw= GW @IP Selection: 2 OracleSBC1 (network-interface)# pri-utility-addr <@IP> OracleSBC1 (network-interface)# sec-utility-addr <@IP> OracleSBC1 (network-interface)# done </pre> |

➤ Wancom1 & Wancom2 physical configuration

| Element | Configuration |
|--------------------------------|---|
| Wancom1 physical configuration | <pre> OracleSBC1# configure terminal OracleSBC1 (configure)# system OracleSBC1 (system)# phy-interface OracleSBC1 (phy-interface)# name wancom1 OracleSBC1 (phy-interface)# operation-type Control OracleSBC1 (phy-interface)# port 1 OracleSBC1 (phy-interface)# slot 0 OracleSBC1 (phy-interface)# wancom-health-score 7 </pre> |
| Wancom2 physical configuration | <pre> OracleSBC1# configure terminal OracleSBC1 (configure)# system OracleSBC1 (system)# phy-interface OracleSBC1 (phy-interface)# name wancom2 OracleSBC1 (phy-interface)# operation-type Control OracleSBC1 (phy-interface)# port 2 OracleSBC1 (phy-interface)# slot 0 OracleSBC1 (phy-interface)# wancom-health-score 8 </pre> |

➤ Wancom1 & Wancom2 network configuration

| Element | Configuration |
|---------|---------------|
|---------|---------------|

| | |
|--------------------------------------|---|
| Wancom1 network configuration | <pre> OracleSBC1# configure terminal OracleSBC1 (configure)# system OracleSBC1 (system)# network-interface OracleSBC1 (network-interface)# name wancom1 OracleSBC1 (network-interface)# pri-utility-addr 169.254.1.1 OracleSBC1 (network-interface)# sec-utility-addr 169.254.1.2 OracleSBC1 (network-interface)# netmask 255.255.255.252 OracleSBC1 (network-interface)# done </pre> |
| Wancom2 network configuration | <pre> OracleSBC1# configure terminal OracleSBC1 (configure)# system OracleSBC1 (system)# network-interface OracleSBC1 (network-interface)# name wancom2 OracleSBC1 (network-interface)# pri-utility-addr 169.254.2.1 OracleSBC1 (network-interface)# sec-utility-addr 169.254.2.2 OracleSBC1 (network-interface)# netmask 255.255.255.252 OracleSBC1 (network-interface)# done </pre> |

➤ Redundancy

| Element | Configuration |
|-------------------|--|
| Redundancy | <pre> OracleSBC1# configure terminal OracleSBC1 (configure)# system OracleSBC1 (system)# redundancy OracleSBC1 (redundancy)# peers OracleSBC1 (rdncy-peer)# name OracleSBC1 OracleSBC1 (rdncy-peer)# type Primary OracleSBC1 (rdncy-peer)# destinations OracleSBC1 (rdncy-peer-dest)# address 169.254.1.1:9090 OracleSBC1 (rdncy-peer-dest)# network-interface wancom1:0 OracleSBC1 (rdncy-peer-dest)# done OracleSBC1 (rdncy-peer-dest)# address 169.254.2.1:9090 OracleSBC1 (rdncy-peer-dest)# network-interface wancom2:0 OracleSBC1 (rdncy-peer-dest)# done OracleSBC1 (rdncy-peer-dest)# exit OracleSBC1 (rdncy-peer)# done OracleSBC1 (rdncy-peer)# name <Target name SBC2> e.g. OracleSBC2 OracleSBC1 (rdncy-peer)# type Secondary OracleSBC1 (rdncy-peer)# destinations OracleSBC1 (rdncy-peer-dest)# address 169.254.1.2:9090 OracleSBC1 (rdncy-peer-dest)# network-interface wancom1:0 OracleSBC1 (rdncy-peer-dest)# done OracleSBC1 (rdncy-peer-dest)# address 169.254.2.2:9090 OracleSBC1 (rdncy-peer-dest)# network-interface wancom2:0 OracleSBC1 (rdncy-peer-dest)# done OracleSBC1 (rdncy-peer-dest)# exit OracleSBC1 (rdncy-peer)# done OracleSBC1 (rdncy-peer)# exit OracleSBC1 (rdncy)# done </pre> |

Reboot primary SBC, redundancy configuration require reboot to take effect:



| Element | Configuration |
|---------|--|
| Reboot | OracleSBC1 # save-config OracleSBC1 # activate-config OracleSBC1 # reboot |

6.4.2 Secondary SBC configuration

Get the wancom0 IP address for the SBC1, try to ping this address on the secondary SBC and acquire its configuration:

| Element | Configuration |
|----------------|--|
| Acquire config | OracleSBC2# ping <Wancom0 SBC1 @IP> OracleSBC2# acquire-config <Wancom0 SBC1 @IP> OracleSBC2# reboot force activate |

Issuing “show health” and “display-current-cfg-version” commands to display the state of redundancy. The same numbers of current configuration version indicate that both SBCs are synchronized.

| Element | Configuration |
|-----------|--|
| SBC state | OracleSBC1# show health OracleSBC1# display-current-cfg-version OracleSBC2# show health OracleSBC2# display-current-cfg-version |

```
Redundancy Protocol Process (v3):
  State           Active
  Health          100
  Lowest Local Address 169.254.1.1:9090
  1 peer(s) on 2 socket(s):
  OracleSBC2: v3, Standby, health=100, max silence=1050
                  last received from 169.254.1.2 on wancom1:0
  Switchover log:
```

```
Redundancy Protocol Process (v3):
  State           Standby
  Health          100
  Lowest Local Address 169.254.1.2:9090
  1 peer(s) on 2 socket(s):
  OracleSBC1: v3, Active, health=100, max silence=1050
                  last received from 169.254.2.1 on wancom2:0
  Switchover log:
```

```
OracleSBC1# display-current-cfg-version
Current configuration version is 57
```

```
OracleSBC2# display-current-cfg-version
Current configuration version is 57
```